Poznań University of Technology Faculty of Computer Science Institute of Computer Science

Bachelor's thesis

# **DESIGN AND IMPLEMENTATION OF COMMUNICATION PROTOCOLS FOR SELF-ORGANIZING MULTI-HOP AD HOC NETWORKS USING XBEEPRO-868 PLATFORM**

Amadeusz Juskowiak, 106453 Tomasz Kuczma, 106548 Mateusz Rybarski, 106572 Maciej Øurad, 106537

Supervisor Professor Jerzy Brzeziński

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> *The scientific man does not aim at an immediate result. He does not expect that his advanced ideas will be readily taken up.*

*His work is like that of the planter — for the future. His duty is to lay the foundation for those who are to come, and point the way.*

# **Abstract**

Wireless ad hoc networks consists of autonomous hosts which have equal status within a network. Unlike managed wireless networks, ad hoc networks are decentralized, there is no pre existing structure. Each host can participate in routing by forwarding packets. Decentralized nature of ad hoc networks makes them more reliable than managed wireless networks, because failure of one host does not cause failure of whole network. This feature makes them also more suitable for some kinds of applications, like sensoric networks.

Message delivery protocols are fundamental for applications that use the network, especially when it comes to sending messages to host that is not directly accessible. Assuming that hosts can be spread over a large area and distance between any pair of adjacent hosts can be large, temporary failures of links and network topology changes should be taken into account. This makes delivery guarantee a hard problem. Another problem is to provide self-organising network. Joining to network should be as simple as possible and maintaining topology should happen in automatic way.

In context of the above observations, in this thesis we propose new communication protocols that provide self-organizing networks and try to ensure delivery of messages between any pair of hosts in a network. Moreover, we provide sample implementation of these protocols. Firstly, motivation and available solutions are shown, then designed protocols are described in Chapter 2. Chapters 3 and 4 contain detailed description of provided implementation of protocols and sample applications respectively. Then, in Chapter 5, different kinds of tests have been carried out.

Test results confirmed that provided solutions and implementation are in line with expectations and assumptions.

# **Contents**





# <span id="page-10-0"></span>**Chapter 1**

# **Introduction**

In recent years there is a growing demand for wireless communication, mostly caused by an increasing number of mobiles devices, such as smartphones, tablets, smart watches and home automation appliances. Lately these devices has got more popular, due to technological advances in hardware that resulted in lower prices and therefore more accessibility. Usually communication between those devices is achieved with the help of *Internet*, to which they connect through a wireless or cellular network. On the other hand direct communication is rarely used, even if a group of devices is in close neighbourhood. This solution has its advantages, such as limited responsibility. The mere thing they are responsible for is handling the communication with a device connecting it to the Internet. However, the massive disadvantage comes from relying on this provided infrastructure, which when fails then no communication can be performed. In order to not depend on any pre-established infrastructure another approach for enabling communication is needed.

One well described solution could be an *ad hoc* network [\[Per08\]](#page-87-0). Ad hoc is a Latin phrase for *"Formed for or concerned with specific purpose"*, *"Improvised and often impromptu"* [\[Dic11\]](#page-86-1). Ad hoc network is a decentralized network, where every device connected to it has the same responsibilities and can act both as a client and server simultaneously. Ad hoc networks can be divided into two categories *single-hop* and *multi-hop*, single-hop implies that the communication between two hosts is done without the help of intermediate hosts, while multi-hop network allows for communication assisted by intermediate hosts. Such a network has a possibility of having a much greater geographical span compared to a single-hop approach, because two hosts do not need to be in transmission range. However, there are substantial consequences of this approach. Now each host is responsible for routing traffic between other hosts. This demands all hosts to posses knowledge about the current network topology, but since it is decentralized network, it brings all the problems that can appear in distributed system.

Another form of comparing different ad hoc networks is to look at how they handle topology changes. A *self-organizing* [\[Poo00\]](#page-87-1) network reacts to those alterations without any assistance, it automatically discovers new hosts willing to join the network as well as drops the hosts that are not responding. These differences are then flooded [\[LK01\]](#page-87-2) throughout the network to ensure that all hosts inside the network are notified about the changes. In this thesis a self-organizing multi-hop ad hoc network is studied. Since ad hoc networks are usually deployed on small wireless devices, they usually have plenty of limitations on resources such as battery power, bandwidth, CPU's computational power and memory capacity. These constraints have to be taken into account, while designing such networks.

As a result of being independent from any pre-established infrastructure, ad hoc networks find many applications. One example is military, which nowadays uses computers extensively in

battlefields. Because of this, features like fault tolerance and on demand setup are of utmost importance, and therefore the use of ad hoc networks for army purposes is widely studied [\[JJV08\]](#page-87-3). It is worth nothing that in case of army application, an important assumption of high mobility has to be included. Ad hoc networks where each node can move without any constraints is known as *MANET* [\[BRG99\]](#page-86-2), an abbreviation for mobile ad hoc network. Another excellent example of application that would greatly benefit from ad hoc approach is post disaster relief. In this case, there is also close to zero infrastructure availability, caused by flood, earthquake or other catastrophic event. Yet another application for instance is wireless sensor network monitoring gas pipes that are spread across a country, where access to cellular network can be limited or impossible. Further adoptions worth mentioning are gaming on portable devices, ubiquitous flea market [\[SNS03\]](#page-88-0) as well as assistance during official meetings. More mobile games nowadays feature multi-player modes allowing to play with friends nearby. With the use of ad hoc network, rather than connecting to a server on the Internet, players can create a single-hop ad hoc network and communicate that way. A ubiquitous flea market is a concept, where users could buy and sell items. As they walk, application would search inside the network for an item match, and upon match request users to make a physical contact. Official meetings or school classes can also take advantage of ad hoc technology. People can easily exchange data within a network, that would be created dynamically just for this purpose. Of course security is something that has to be considered. Finally, Internet of Things *(IoT)* is a raising area of research. Its intention is to create a massive network of inter connected devices using existing infrastructure such as Internet. However, over the last decade use ad hoc networks was investigated to supplement the Internet [\[RTB](#page-87-4)+13]. The word *Things* from the definition refers to broad range of devices, including heart monitoring implants, energy management sensing and actuation systems, home automation systems and environment monitoring systems. These are just a few examples from a wide group of real world applications.

When studying currently available products based on ad hoc network, it can be observed that the market is still in its infancy. Moreover, there exist a number of available solutions to different problems that rely on existing infrastructure. Combining those two facts allows for creating a product, which would not only innovate, but could also compete with a lower price. One example of it, would be a modern timing system used in different sport events, such a marathons and triathlons. Main task of this system is to exchange data between points placed within the race route. These points would collect timestamps of each participant passing by. Data would be aggregated in real-time at a single node in order to calculate intermediate and lap times as well as reveal any cheaters. In already existing solutions data is sent by GPRS modules (for an extra fee paid to GSM operator). Therefore, not only it depends on the infrastructure, but also generates more costs associated with data transfers. Developing a modern timing system built with ad hoc network as its base for communication, would therefore be more resistant to failures and additional redundancy would increase this resistance even further. Data transfer costs can also be removed depending on the type of communication used between two adjacent nodes. Using a publicly available frequency band is an example of this. There is currently no timing system on the market, which is based on ad hoc approach. The reason for this is mostly due to the fact that creating such product comes with more responsibilities. Finally, it is the software or hardware of the device that is responsible for routing packets throughout the network, since there is no existing infrastructure. Moreover, ad hoc networks face many other problems such as security and manufacturing a final product that clients would be happy with is not an easy task. However, for a timing system, weaker assumptions can be made, which greatly simplify the design of the network, for instance: predefined shared key required to join network and no network mobility.

That being said, motivation for this thesis came precisely from observing the market of cur-

rent timing systems and noticing that an ad hoc approach could seriously disrupt it. Therefore, the goal of this thesis is to design and implement communication protocols that implement selforganizing multi-hop long distance ad hoc network. Long distance means that it allows for a long distance communication between two adjacent nodes. Together with multi-hop feature of the network, it implies that the network can have an immense geographical span. Consequently, these protocols would fit extremely well as a core of a modern timing system used in sport competitions, where gap between two points can be large. In order to support long range communication, radio modules from Digi International [\[Dig15\]](#page-86-3), namely XBeePro series were chosen as a base for transmission. There are two available models, which operate on two different frequencies. First one being XBeePro-868, that uses 868MHz frequency and the second, XBeePro-900 that works on 900MHz frequency. The latter one implements a proprietary mesh network protocol, where each node acts as a router [\[JS03\]](#page-87-5). However, in Europe there is legal problem. European Union directive 87/372/EWG [\[Eur09,](#page-87-6) [Eur11\]](#page-87-7) does not allow public access on 900MHz frequency. In that case to create mesh network using XBeePro in Europe it is necessary to implement own multi-hop ad hoc network and use XBeePro-868 [\[Dig14b\]](#page-86-4).

Knowing the hardware, the network is going to operate on, following assumptions were made in order to reduce complexity of the project. No network mobility, which otherwise would complicate routing and discovery protocols. Security of the network that relies on the hardware and users, meaning XBeePro-868 modules are responsible for encryption and that only authorized people know the password. Therefore, this network could not be used for applications that involve public access, because the protocol does not solve typical security problems such as Man-in-the-Middle and Denial of Service attacks [\[HBC01\]](#page-87-8). Another important assumption concerns network partitions. Partition occurs if during operation of network there exist two nodes without a path. In the design of the protocol it is assumed that permanent network partition cannot arise.

After determining assumptions, these functional and non-functional requirements were defined to ensure that the designed protocols could be used in different applications, for instance previously described timing system or for that matter any data collecting system, that would have static nodes (node with no mobility). First important functional requirement is that system shall handle topology changes. Meaning new nodes joining the network shall get notified about current network topology and vice versa. It also implies that temporary failures of nodes or links will be noticed and propagated throughout the network as well. Second functional requirement concerns packet delivery, and it states that the protocol shall deliver a packet sent from node *A* to node *B* eventually. However, it can deliver the same packet more than once. On the other hand non-functional requirements are qualities of the system. One requirement for the implementation of the protocol is to be portable for all  $\text{POSIX}^1$  $\text{POSIX}^1$  compliant operating systems with access to essential dependencies. Second quality requirement is to abstract and hide routing from the end user of network. Another non-functional requirement, which is partially coupled with the previous one, is to provide an API[2](#page-12-1) of end-point to end-point packet transmission to the end user.

The thesis was written with the following structure. In Chapter 2, general project concept is discussed. Starting with the description of data types and variables that the protocol is using. Then each part of the protocol is explained. Consisting of network topology maintenance and discovery, as well as packet routing and mechanism that try to ensure packet delivery. The chapter ends with proof of correctness for the described protocol. Chapter 3 contains a sample implementation of the protocol discussed in the previous chapter. It begins with detailed explanation of the

<span id="page-12-0"></span><sup>1</sup>POSIX is abbreviation for Portable Operating System Interface and is an operating system compatibility standard

<span id="page-12-1"></span><sup>2</sup>API stands for Application Programming Interface

chosen hardware that is directly responsible for radio communication. It also shows the control system platform that actually runs the implementation, giving a rundown of multiple choices for such control system. Next, the implementation itself is described. Exact modules responsible for different tasks such as communication with the radio hardware, providing API for the user, routing itself are illustrated. After simulator used for performance analysis is also described. At the end testing methodology applied during development process is also explained. Next, Chapter 4 discusses sample applications of the implemented protocol using provided API. It shows 3 dierent applications: echo, temperature reading and console together with detailed implementations. Finally, other possible areas of usage are mentioned. Chapter 5 focuses on evaluating performance of the protocol implementation with regards to network size, topology and environment. It starts with assessing complexity of topology discovery process. Afterwards, load and spike testing is described and performed. Then, harsh environment testing is conducted. Finally, analysis of all results is carried out. Eventually, summary of the study is discussed in Chapter 6.

In this work, specific tasks were divided among the group. Implementation of the protocol was assigned to three people: Tomasz Kuczma, Mateusz Rybarski and Maciej Øurad. The design of the protocol was also split into parts. Delivery guarantee was done by Tomasz Kuczma and Mateusz Rybarski, while discovery and maintenance of topology was designed by Amadeusz Juskowiak. Routing of packets was developed by Amadeusz Juskowiak. Last part of the protocol, proof of correctness was completed by Maciej Øurad. Network simulator and its further integration with various tools was done by Amadeusz Juskowiak and Mateusz Rybarski. While integration, performance and reliability tests together with fixtures were conducted by Tomasz Kuczma. Software engineering issues and unit testing were solved by Tomasz Kuczma and Maciej Øurad. Finally, implementation of sample applications was done by Mateusz Rybarski, however the driver for the end user was designed and implemented by Amadeusz Juskowiak.

# <span id="page-14-0"></span>**Chapter 2**

# **Project concept**

In following chapter, general concept of communication protocols provided in this thesis is described. Then correctness of various properties of these protocols are proven. In context of protocols and algorithms operating with network, there is a need to choose network representation used in this thesis. Because of nature of designed protocols, where in every pair of adjacent nodes, both of them have to see each other to make communication possible, undirected graph  $G = (V, E)$  is chosen. This graph consists of a set of nodes *V* and a set of edges *E*.

# <span id="page-14-1"></span>**2.1 Basis of operation**

This section describes basis of operation and algorithms used in designed protocols. Following subsections show how topology of the network is discovered and maintained, how packets with data sent by user applications are routed within the network and how the protocols try to ensure delivery of these packets. Firstly, there is a need to define data types, operations and events used in designed protocols.

### <span id="page-14-2"></span>**2.1.1 Data types, variables, operations and events**

Following operations, events and data types are used in designed protocols.

### **Data types**

Pseudocode [1](#page-15-0) shows data types used in protocols. It is worth to notice, that *data* field of *DATA* packet type is not fully defined here. This type depends on implementation details, so its full definition is omitted here. Different implementations can send via network different types of data.

## **Variables**

At every node, variables from pseudocode [2](#page-15-1) are defined. self represents node itself. In addition, every node contains graph that represents its knowledge about the state of the network. In the rest of this chapter, it will be named *local network*. Inside pseudocodes, *G* stands for whole local network, *V* stands for a set of known nodes, *E* stands for a set of edges known to given node.

### **Operations**

Operations that take argument of type *PACKET* can also take argument of any type that extends *PACKET*.

• add\_node(node : NODE) – adds node to graph that represents local network,

**Pseudocode 1:** Data types used in protocols

<span id="page-15-0"></span>**type**  $NODE$ : Integer;  $\overline{\phantom{a}}$  /\* Unique identifier of node \*/ **type** *EDGE* **is** *array[1..2]* **of** *Node*; /\* Represents edge between two nodes \*/ **type** *EDGE PARAMETERS* **is record of** good : Integer;  $\frac{1}{2}$  Integer;  $\frac{1}{2}$  Parameters of given edge  $\frac{1}{2}$ retries : Integer; errors : Integer; **end type** *PATH* **is** *list* **of** *NODE* ; /\* The path of a packet \*/ /\* types of packets sent between nodes \*/ **type** *PACKET* **is record of** ; /\* Base type of packets used in algorithms \*/ **end type** *NODE BROADCAST* **extends** *PACKET* **is record of** | node : NODE; **end type** *GRAPH* **extends** *PACKET* **is record of** edges : list **of** EDGE; **end type** *EDGE DROP* **extends** *PACKET* **is record of** edge : EDGE; **end type** *DATA* **extends** *PACKET* **is record of** id : Integer;  $\sqrt{*}$  unique identifier of packet  $*/$ source : NODE; destination : NODE; data : ... ;  $\sqrt{*}$  data sent between nodes  $*/$ visited : list **of** NODE; **end type** *ACK* **extends** *PACKET* **is record of** id : Integer;  $\frac{1}{2}$  /\* identifier of corresponding DATA packet \*/ status : Integer; path : PATH; params : list **of** *EDGE PARAMETERS* **end**

**Pseudocode 2:** Variables defined at every node

<span id="page-15-1"></span>self : NODE;

packet history : map of (DATA, timestamp);

- remove\_node(node : NODE) removes node from graph that represents local network,
- add\_edge(a : NODE, b : NODE) adds edge from a to b to graph that represents local network,
- add edge(edge : EDGE) adds edge to graph that represents local network,
- remove\_edge(edge : EDGE) removes edge from graph that represents local network,
- calculate antireliability(G) calculates reliability of edges in given graph, described in packet routing section,
- *•* path(source : NODE, destination : NODE) : PATH obtains and returns shortest path from source to destination in graph *G*,
- calculate\_timeout(path : PATH) : Timestamp calculates and returns timeout for acknowledgement packet for given path,
- history remove(id : Integer) finds and removes from packet history entry for packet with given id,
- *•* history\_find(id : Integer) : DATA finds in packet history and returns entry for packet with given id,
- *•* deliver(receiver : NODE, packet : DATA) delivers packet of type *DATA* from current node to node Receiver
- receive(sender : NODE, packet : PACKET) receives packet at current node from node Sender
- send(receiver : NODE, packet : PACKET) sends packet from current node to adjacent node Receiver,
- broadcast(packet : PACKET) sends packet from current node to every adjacent node.
- error (message: String) reports error to application, used only when there is no route to destination,
- *•* process(data) process data by application.

### **Events**

Events that take argument of type *PACKET* can also take argument of any type that extends *PACKET*.

- e\_timeout\_expires(packet : DATA) occurs when timeout for given *DATA* packet expires,
- *•* e\_antireliability\_exceeds(edge : EDGE) occurs when antireliability for given edge becomes too high,
- e receive(sender : NODE, packet : PACKET) occurs when receive operation is invoked at current node,
- *•* e\_deliver(receiver : NODE, packet : DATA) occurs when deliver operation is invoked at current node.

### <span id="page-16-0"></span>**2.1.2 Network topology discovery**

To provide self-organizing network, topology has to be discovered at the beginning of operation. Solving this problem is divided into two steps:

- *•* Informing visible nodes that new node wants to join the network,
- *•* Informing rest of the network about new node and edges in the network.

To join the network, new node broadcasts packet of type *NODE BROADCAST*. This packet contains identifier of broadcasting node that is unique. If any other node receives this packet, e\_receive(sender : NODE, packet : NODE\_BROADCAST) event occurs and procedure from pseudocode [3](#page-17-1) is executed on the receiving node. Node received in this packet and edge from this node to itself is added, unless existing in the graph that represents local knowledge of the network. If

**Pseudocode 3:** Receiving NODE BROADCAST packet event handler

<span id="page-17-1"></span>

	when $e\_receive(sender : NODE, packet : NODE\_BROADCAST)$ :
	if $node \notin V$ then
	add_node(packet.node);
end	
	if $\{self, packet node\} \notin E$ then add_edge(self, packet.node);
	$p_{\text{graph}}$ : GRAPH; $p_{\text{-}graph.nodes} := get_{\text{-}edges}($ $broadcast(p_graph);$
end	
end	

the edge was added, every edge in the graph is broadcasted using packet of type *GRAPH*. *GRAPH* packet contains information about every edge that is known to broadcasting node. When packet of type *GRAPH* is received, e receive(sender : NODE, packet : GRAPH) occurs and procedure from pseudocode [4](#page-17-2) is executed. Edges from the packet are merged with local network graph. If it has changed, new packet of type *GRAPH* is broadcasted with current graph.

**Pseudocode 4:** Receiving GRAPH packet event handler

```
when e receive(sender : NODE, packet : GRAPH):
   change : Bool;
   change := False;foreach edge : packet.edges do
       if edge \notin E then
          add edge(edge);
          change := True;end
   end
   if change = True then
       p graph : GRAPH;
       p_{\text{graph}.\text{nodes}} := get_{\text{edges}}(broadcast(p_graph);end
end
```
# <span id="page-17-0"></span>**2.1.3 Network topology maintenance**

After discovering topology, it is needed to ensure that nodes are still able to deliver packets via edges in graph and nodes are still working. This is done in two steps:

- Removing faulty edges from graph,
- *•* Rediscovering nodes and edges in case of temporary breakdown.

Firstly, when antireliability measure for given edge exceeds threshold value (antireliability measure is described widely in Packet routing section), e\_antireliability\_exceeds(edge : EDGE) event occurs and code from procedure shown in pseudocode [5](#page-18-1) is executed. Packet of type *EDGE DROP* is broadcasted to inform other nodes, that given edge should be removed from graph because there is a possibility undelivering packet using this edge.



```
when e antireliability exceeds(edge : EDGE):
   p edge drop : EDGE DROP;
   p-edge-drop.edge := edge;
   broadcast(p\_edge\_drop);end
```
When packet of type  $EDGEDROP$  is received, e\_receive(sender : NODE, packet : EDGE\_DROP) event occurs (pseudocode [6\)](#page-18-2). The received edge is removed from the graph. If it was removed successfully, the *EDGE DROP* packet is broadcasted again from current node. Otherwise, edge was deleted earlier or have never existed for this node and *EDGE DROP* packet is not broadcasted. This operation allows to remove faulty edges in whole network. If node without any edge exists in the network, it should be removed locally from graph.

**Pseudocode 6:** Receiving EDGE DROP packet event handler

<span id="page-18-2"></span>

Secondly, once in a while, from every node packet of type *NODE BROADCAST* is broadcasted and procedure similar to topology discovering is happening in the network. This allows rediscovering nodes and edges if they were removed from graph in case of broadcasting *EDGE DROP* packet.

# <span id="page-18-0"></span>**2.1.4 Packet routing**

As mentioned at the beginning of this chapter, network is represented by undirected weighted graph. Packet routing determines shortest path in this graph, from packet source node to packet destination node. Important thing is that following rules are applied only for packets of type *DATA*. These packets contains addresses of source and destination nodes, data that needs to be delivered and list of visited nodes by packet.

Following subsections describe how antireliability measure used as weights on edges is calculated, how packet is delivered to destination node and how parameters used to calculate antireliability measure on edges are updated with acknowledgement packets.

### **Antireliability measure**

Network is represented by undirected weighted graph, where antireliability measure acts as edge weight. This measure is a real number greater than or equal to 0. Greater the measure is, reliability

of given edge is lower.

Antireliability measure evaluation is based on parameters stored for every edge in the graph:

- **GOOD** which determines number of delivered packets using given edge,
- **b** ERRORS which determines number of undelivered packets using given edge,
- **betwines and determines number of retries of packets on given edge.** *RETRIES*

parameter *RETRIES* parameter makes *ERRORS* It is worth to notice, that every increase of is easy of *Am Souther is a performal of 1.8 updatines word and antireliation* and  $\alpha$  is  $\alpha$  and  $\alpha$  is  $\alpha$ calculated:

$$
d_{\text{WE}} = \frac{1 + 4000}{1 + 40000} = 0.1
$$

### **Packet delivery**

<span id="page-19-1"></span>packet from source node to destination node, an antireliability measure is *DATA* In order to send calculated for every edge in the graph as described above. Then, the path from source node to vilidsilərina dirm [9diQ] mahrogla draq isətroda a'cırladi man bujub antireliyya batım alability measures as edge weights. Subsequently,  $D X Y Y$  packet is sent to first node on obtained path. There is possibility, that path obtained will be empty. That means destination is inaccessible from current node. If such situation occurs, proper error should be raised to application layer. This . [7](#page-19-1) process is shown in pseudocode

```
Pseudocode 7: e_deliver event handler
: deliver(receiver : NODE, packet : DATA) e when
                                           calculate_antireliability(G);
                                                                        path : PATH;
                                          path := path(self, receiver);
                                                                theory \emptyset \neq \emptyset if \emptyset if
                                           senq(bath.first, packet);
                                                             t : \text{Im}(\text{div}) \to \text{Im}(\text{div})t := \text{cals} calculate_timeout(path);
                                 \mathbf{u} \cdot \mathbf{v} = \mathbf{v} \cdot \mathbf{v} + \mathbf{v} \cdot \mathhistory.add(packet, t);
                                                                                     end
                                                                                            else
                                    theory if \phi \neq p is parameter in \phib-ack : ACK;
                                        \alphaid: id\alphaexek.id; packet.id;
                                            : self; proper : = self; :p_ack.path := packet.visited;
                         aearq(b serb.path.last, p sek):
                                                                                     else
                \mathcal{L}(\mathbb{Q}_{\mu})error\mathbb{Q}_{\mu}o route to destination");
                                                                                     end
                                                                                            end
                                                                                                   end
```
mis received on intermediate node, and *particle node*, list of visited nodes stored in *DATA* When packet of two types  $\mu$ packet is updated with current node, antireliability measures are evaluated at current node and path from current node to destination node is obtained, trying to avoid already visited nodes. *DATA* Afterwards, packet with data is sent to first new to the sense of the new reads of the part of the DTA  $\alpha$ 

reaches its destination, acknowledgement packet is sent back to the source node and received data can be processed by node. This process is presented in pseudocode [8.](#page-20-1)

**Pseudocode 8:** Receive DATA packet event handler

```
when e receive(sender : NODE, packet : DATA):
   if self = packet.destination then
      p ack : ACK;
      p_ack.id := packet.id;p_ack.status := 0;
      p_{ack}.path := packet.visited;send(p_ack.pathu, p_ack);process(packet.data);
   else
      packet.visited.add(self);
      deliver(self, packet.destination, packet);
   end
end
```
## **Acknowledgement packets**

Acknowledgement packets (packets of type *ACK*) are sent back from destination node to source node, by exactly the same path the corresponding packet was delivered to the destination node. The *ACK* packet contains parameters of every edge on the path of corresponding *DATA* packet, which are used to update graph on intermediate and source nodes. Process of updating these parameters strongly depends on implementation details and used hardware, so it is not widely described in this chapter. If *DATA* packet could not be delivered, *ACK* packet contains also information about last node which could not deliver the packet.

# <span id="page-20-0"></span>**2.1.5 Ensuring delivery**

One of main goals of this project is to try to ensure that *DATA* packet sent from source node will be delivered to destination node or message with error will be send to application when packet could not be delivered. As it is mentioned in assumptions section of first chapter, it is assumed that packet will be delivered to its destination node, but there are cases when packet can be delivered more than once. To aim this goal, two mechanisms are used: acknowledgement packets and timeouts. For purpose of delivery guarantee, there is a need to remember at every node, every packet that was sent or was transmitted . This is remembered in packet history.

### **Acknowledgement packets**

Acknowledgement packets are very important to delivery process. After *DATA* packet reaches its destination, *ACK* packet is sent to *DATA* packet source node by exactly the same way corresponding *DATA* packet was transmitted. When *ACK* packet reaches corresponding *DATA* packet source node, this node knows that packet was delivered successfully. As it is mentioned in previous subsection, acknowledgement packets are also used when *DATA* packet can not be delivered. In this case, *ACK* packets contains information about node that could not deliver *DATA* packet to next node. When source node receives *ACK* packet with such information, it knows that delivery of *DATA* packet failed and this packet should be retransmitted (probably with different path, if it is possible). Whole procedure is shown on pseudocode [9.](#page-21-0)

**Pseudocode 9:** Receive ACK packet event handler

<span id="page-21-0"></span>

It should be noticed that when source node of corresponding *DATA* packet receives *ACK* packet, it should remove entry about *DATA* packet from packet\_history. There are cases when *ACK* packets can not reach source node of corresponding *DATA* packet. These situations are resolved using timeouts mechanism described below.

### **Timeouts**

<span id="page-21-1"></span>As mentioned, timeouts mechanism repeats delivery process of *DATA* packet when corresponding acknowledgement packet does not reach source node. For every outgoing packet, receipt time of corresponding acknowledge packet is calculated. Equation [2.2](#page-21-1) shows how this time is calculated:

$$
T = 2c \cdot \left( \sum_{e \in E_p} \left( d_e \cdot \left( 1 + \frac{RETRIES_e}{max\{GOOD_e - RETRIES_e, 1\}} \right) \right) + \sum_{v \in V_p} t_v \right) \tag{2.2}
$$

Where:

*RET RIES***,** *GOOD* are parameters of given edge, described in antireliability measure subsection,

- $E_p$  is a set of edges in packet path,
- *d<sup>e</sup>* is a delay on given edge,
- *V<sup>p</sup>* is a set of nodes in packet path,
- $t<sub>v</sub>$  is a time of processing single packet on given node
- *c* is a constant greater or equal 1.

Timeout is calculated at source node of *DATA* packet and stored in packet\_history. Action, which takes place after timeout expires is shown on pseudocode [10.](#page-22-3) After timeout expires, *DATA* packet is retransmitted (possibly with a different path) and new timeout is calculated.

```
Pseudocode 10: e_timeout_expires event handler
```

```
when e timeout expires(packet : DATA):
   history_remove(packet.id);
   deliver(packet.destination, packet);
end
```
# <span id="page-22-0"></span>**2.2 Proof of correctness**

The following section's plan is to show what correctness of a protocol is and what is required to prove such thing.

### <span id="page-22-1"></span>**2.2.1 Assumptions**

This section focuses on presenting assumptions about network and the environment that surrounds it. Furthermore, based on these assumptions and special cases of the protocol that can appear, it will be clear which properties the protocol has.

Critical assumptions for understanding the protocol are:

- *fail-stop* node [\[SS83\]](#page-88-1) means that the process (in this case a node) when fails, will go into halt mode instead of executing erroneous state. This also indicates that other processes will be able to detect that and deal with it by not trying to communicate. Fail-stop also implies that a process will lose all state after failure, therefore it is equivalent to introducing a new node.
- *• non-permanent network partitions* is a very important assumption for the correctness of the protocol. It promises that if a network partition occurs, it will not be perpetual. A network partition happens when a failure of a single node leads to a situation, where there exist two correct nodes unable to communicate with each other.
- *changing topology* one way of permanently changing topology is to introduce new nodes. However, this does not provide any bad consequences unlike permanent failures, which would highly complicate the proof of correctness.
- *perfect links* [\[CGR11\]](#page-86-6) every two adjacent nodes are in a perfect link. Meaning that if the nodes are correct (neither sender nor receiver is down), then if one process sends a message to its neighbour, neighbour will obtain it eventually. This property holds because of the nature of XBeePro-868 hardware, which handles retransmits by itself, and if the receiver node is down then it returns an error status. It also implies no duplication of messages between those two nodes and a requirement of an originator of message. No message can be received without it being sent first.

# <span id="page-22-2"></span>**2.2.2 Special cases**

The protocol uses acknowledgments and timeouts as a way of telling the originator of message if it has been successfully or unsuccessfully delivered. Because of this, there is a situation, which makes it impossible to prove that the protocol has a *Reliable Delivery* property [\[CGR11\]](#page-86-6). This property says that if a correct process requests to send a message *m* to a correct process *q*, then *p* eventually delivers *m* to *q*.

The situation that causes this property to fail is depicted in the figure [2.1](#page-23-1) and is the following: Process *p* sends a message *m* and correctly delivers it to process *q*, after receiving the message

<span id="page-23-1"></span>

Figure 2.1: ACK returns undelivered, after delivering message

process *q* sends back the acknowledgment message *ACK* to process *p*. While the *ACK* is being routed to originator, the network is temporary partitioned, leaving the *ACK* message without a path to originator. This after some time *t* causes timeout to expire and originator attempts to retransmit the message again. However, the network stays partitioned and eventually originator reports that the message has been unsuccessfully delivered. This shows that the reliable delivery property does not hold, since reporting undelivered at the originator does not guarantee anything. Receiving acknowledgment without error status means of course that the message has been successfully delivered, but that is not enough for proving *Reliable Delivery*. Following logical statements [2.3](#page-23-2) show this property. It can be seen that receiving status *OK* implies that message was delivered. However, delivering the message does not imply that status *OK* will be received. Same case applies to receiving status *NOT OK*, which doesn't imply that message was not delivered.

<span id="page-23-2"></span>
$$
OK \rightarrow Delivered,
$$
  
¬(Delivered → OK),  
¬Delivered → NOT\_OK,  
¬(NOT\_OK → ¬Delivered),  
□ (2.3)

### <span id="page-23-0"></span>**2.2.3 Proving correctness of the protocol**

To consider any distributed protocol correct requires two classes: *safety* and *progress* to be satisfied [\[Lam77,](#page-87-9) [ADS86\]](#page-86-7). Safety properties enforce that something cannot happen. They are used to prevent anything bad from happening. However, only specifying safety properties is not enough, a trivial program that does not do anything would count as correct. Therefore the second class of progress properties is needed. Progress properties (also known in literature as *liveness* properties [\[CGR11\]](#page-86-6)) are there to ensure that something good occurs. Moreover, these properties state that for any time *t*, they will be fulfilled at time  $t' > t$ .

The properties of protocol are defined in [2.1.](#page-24-0) Immediately it can be observed that *Stubborn delivery* is a *progress* property and *No creation* is a *safety* property. Stubborn delivery is weaker

#### <span id="page-24-0"></span>**Module:**

**Name:** StubbornPointToPointLinks, **instance** *sl*.

### **Events:**

**Request:**  $\langle sl, Send \mid q, m \rangle$ : Requests to send message *m* to process *q*. **Indication:**  $\langle sl, Deliver | p, m \rangle$ : Delivers message *m* sent by process *p*.

#### **Properties:**

**SL1:** *Stubborn delivery:* If a correct process *p* sends a message *m* once to a correct process *q*, then *q* delivers *m* an infinite number of times.

Table 2.1: Interface and properties of Stubborn point to point links

<span id="page-24-1"></span>**SL2:** *No creation:* If some process *q* delivers a message *m* with a sender *p*, then *m* was previously sent to *q* by process *p*.



Figure 2.2: Stubborn Links time graph

of course than most progress properties, most notably reliable delivery. It assures that if and only if the message *m* is transmitted infinite number of times from a process *p*, then the process *q* will receive it an infinite number of times. Figure [2.2](#page-24-1) shows a time line of the same frame going through the network, and either delivering or not delivering.

**Theorem 1.** *If two presented properties* Stubborn delivery *and* No creation *are met, then the whole protocol implements Stubborn Point to Point links interface.*

*Correctness Arguments.* It is trivial to prove that the *No creation* property holds. Previous assumption that the network is operating on perfect links, makes it very easy to say that this property cannot be violated, since perfect links are much stronger property than e.g. fair-loss links [\[CGR11\]](#page-86-6) which already cover *No creation* property. The progress property on the other hand, requires to look at the protocol and assumptions. Because Stubborn delivery sends a message

infinite number of times, then the receiver will get also infinite number of messages. The actual difference in number might be huge between sent and received messages, but looking at infinity it holds. Assumption about temporary network-partitions is very important, because retransmitting messages infinite number of times, doesn't assure the receiver that he is also going to get infinite number of messages. In fact if the partitions could be permanent, the property would not hold. Obviously the protocol is not efficient if the message was received and the other side doesn't need it anymore. Therefore, the implementation of the protocol sends acknowledgment packets, which stop sending the same message. This is merely an implementation detail, so it can be done. $\Box$ 

# <span id="page-26-0"></span>**Chapter 3**

# **Project**

Algorithm implementation is required to study its behaviour in real life situations and perform various tests. Two distinct implementation sections can be designated – hardware-related design and software implementation. Software implementation requires connection to physical radio device or simulator environment.

# <span id="page-26-1"></span>**3.1 Infrastructure**

Proposed algorithm does not require any specific hardware platform, however its design process was inspired by XBeePro-868 devices produced by Digi International Inc. Although XBeePro-868 device provides support for point-to-point and point-to-multipoint, implementation of mesh networking is omitted in the current radio firmware.

XBee-Pro 868 provides physical, mac and security layers upon which our network layer is built. The radio module connects with controller (such as personal computer, embedded system or microcontroller) using industry standard serial communication. Therefore communication in heterogeneous computer environment is possible. It should be noticed that line-of-sight distance between two adjacent nodes can be as long as 40 kilometers with proper antennas.

## <span id="page-26-2"></span>**3.1.1 XBeePro-868 radio module**

Digi XBeePro-868 radio module is complete radio solution for point-to-point and point-to-multipoint communication. Essential components such as UART transceiver, radio modem and amplifier were embedded into small footprint (figure [3.1\)](#page-27-0) module. Complete module documentation is available in manufacturer provided manual [\[Dig14b\]](#page-86-4).

Radio amplifier transmits signal with 315mW power, which combined with built-in high signalto-noise receiver and proper antenna allows to create stable link over 40km line-of-sight distance.

### **Specification**

XBeePro-868 module requires 3.3V power source capable of delivering current up to 500mA. Builtin UART allows variety of data rates (1200 - 230400bps) to be used with communication to controller. Overview of the parameters is available in table [3.1,](#page-27-1) however complete specification is available in the datasheet [\[Dig11a\]](#page-86-8).

Modules implement three networking layers – physical baseband layer, mac layer and security layer. Modules are addressed with 64-bit MAC address set by the manufacturer. Beside that nodes need to be within same network (determined by user configurable network identifier). Transmissions are encrypted using symmetrical AES-128 algorithm [\[DR91\]](#page-86-9) (its support is built-in

<span id="page-27-0"></span>

Figure 3.1: XBeePro-868 mechanical drawing [\[Dig11a\]](#page-86-8)

Table 3.1: XBeePro-868 specification

<span id="page-27-1"></span>

Parameter	Value
Data rate	24Kbps
Indoor range	Up to 500m
Outdoor range	Up to 40km
Transmit power	1mW (0dBm) to $315mW$ (+25dBm)
Receiver sensitivity	$-112dBm$
Frequency band	868 MHz (SRD g3 Band 869.525MHz)
Addressing	64-bit MAC address, 16-bit Network ID
Encryption	128-bit AES

into hardware and transparent to user). Therefore knowledge of network identifier, mac addresses and encryption key is required to transfer and receive messages with these devices, making it quite secure solution.

<span id="page-27-2"></span>Interfacing with other electronics is conducted using 3.3V CMOS level pins. The pinout is described in table [3.2.](#page-27-2)

Pin number	Type	Function		
1	Power	3.3V power supply		
$\overline{2}$	Output	UART data output		
3	Input	UART data input		
4	Input/Output	GPIO		
5	Input	Reset trigger		
6	Input/Output	GPIO, PWM or RX signal strength indicator		
7	Input/Output	GPIO or PWM		
8	Not connected	None		
9	Input/Output	GPIO or sleep trigger		
10	Power	Ground		
11	Input/Output	GPIO or analog input		
12	Input/Output	GPIO or serial CTS signal		
13	Output	GPIO or sleep indicator		
14	Power	Voltage reference for analog inputs		
15	Input/Output	GPIO or connection indicator		
16	Input/Output	GPIO or serial RTS signal		
17	Input/Output	GPIO or analog input		
18	Input/Output	GPIO or analog input		
19	Input/Output	GPIO or analog input		
20	Input/Output	GPIO or analog input		

Table 3.2: XBeePro-868 pinout

<span id="page-28-2"></span><span id="page-28-0"></span>

Table 3.3: Radio power levels

(a) RP-SMA connector (b) U.FL connector (c) Wire antenna

Figure 3.2: Available XBeePro-868 variants[\[Dig14a\]](#page-86-10)

#### **Radio parameters**

Radio transmitter is capable of transmitting signal up to 315mW (+25dBm). Its power can be limited with 5 discrete levels presented in table [3.3.](#page-28-0) Receiver is rated with -112dBm sensitivity. Radio works in 868MHz frequency band (in specific SRD g3 Band 869.525MHz) which is available in Europe and its usage is regulated by ETSI standard [\[Eur11\]](#page-87-7) and EU directive [\[Eur09\]](#page-87-6).

The module is capable to transfer data with 24Kbps data rate, however its duty cycle is limited to  $10\%$  $10\%$ <sup>1</sup>. Duty cycle causes effective throughput to be limited to 2.4Kbps. Duty cycle implementation by Digi allows to send data for 6 minutes in total in each hour. For remaining 54 minutes device is in passive mode – it can only receive frames as transmitting is disabled.

Modules are available with various antenna connectors – RP-SMA connector (figure [3.2a\)](#page-28-2), U.FL connector (figure [3.2b\)](#page-28-2) or built-in wire antenna (figure [3.2c\)](#page-28-2). In total 5 devices were used during development, four devices with RP-SMA connectors and one device with built-in wire antenna.

RP-SMA connector is more reliable than U.FL, as U.FL was designed to connect small embedded antennas, while RP-SMA is industry acceptable external low-power antenna connector.

Tests were conducted using small +2dB omnidirectional antennas (figure [3.3\)](#page-29-0). For large distance communication high gain antennas are required. In this case antenna specification should be selected depending on environment and required distance. Such selection and appropriate calculations and simulations are not part of the protocols concept.

### **Communication methods**

XBeePro-868 requires at least four pins (power supply, ground, data input, data output) connected to make UART connection available. Parameters of the connection such as data rate, parity, stop bytes and flow control must be the same on the module and its controller.

Two modes of communication are available – transparent mode and API mode.

#### **Transparent mode**

<span id="page-28-1"></span> $1$ Duty cycle is limited by ETSI standard. [\[Eur11\]](#page-87-7)

<span id="page-29-0"></span>

Figure 3.3: Omnidirectional +2dB 868MHz antenna with RP-SMA connector[\[Fac15\]](#page-87-10)

In transparent mode any bytes sent to data input pin are encapsulated automatically into frame and sent to other XBee device. When any data are received they are decapsulated and sent to the controller using data output pin. Therefore such two devices emulate direct serial port connection.

Devices must be configured in order to use this mode – two devices must be paired together.

## **API mode**

API mode allows for more advanced transmissions. Controller is required to provide correct frames into data input pin. When data are received, frame is not decapsulated – raw frame is sent out using data output pin. Other frames can be generated internally and sent by XBee device as response to various events.

In this mode programmer can explicitly specify destination of a frame. In similar fashion received frames contain explicitly source mac address. This behavior is practical and it is used in software implementation of the protocols concept.

### **Configuration**

XBeePro-868 modules are configured using AT-inspired command set.

Commands can be entered in various ways. AT Command Mode can be entered using following algorithm – wait one second, enter +++, wait one second. In this mode settings can be managed using AT syntax [\[Dig14b,](#page-86-4) p. 16]. When module is in API mode, AT command can be encapsulated into command frame [\[Dig14b,](#page-86-4) p. 38, 43]. It is possible to execute AT command on remote XBeePro-868 module from another reachable device using remote command frame [\[Dig14b,](#page-86-4) p. 42-43]. Entering AT Command Mode in API mode is still possible using the same algorithm.

Values are stored in built-in reprogrammable EEPROM memory. To replace stored values command ATWR is required.

Assortment of parameters are configurable – comprehensive list of settings and associated AT commands is available in the manual [\[Dig14b,](#page-86-4) p. 27-34]. Some of parameters are read only, an essential one is mac address which can be read using ATSH and ATSL commands.

In order to make use devices as radio transceiver for implementation of thesis algorithm some settings must have been changed – list of settings changed for each device is available in table [3.4.](#page-30-1)

<span id="page-30-1"></span>

AT command	Default value	New value	Description
AΡ	0x00	0x01	Enables API mode
BD	0x03	0x07	115200bps baud rate
ΝI	, ,	various	Node name
EE	0x00	0x01	Enables AES encryption
ΚY	unknown	'HelloWorld'	Encryption key
PL	0x04	depends on test	Power level (see table 3.3)

Table 3.4: List of changed settings

## <span id="page-30-0"></span>**3.1.2 Control system**

XBee868-Pro can not be completely functional without control device connected to the module using UART connection. Control system is platform for executing software implementation of the routing algorithm. Radio module can be connected directly to the control system using UART or using some sort of adapter if UART is not available on target. This allows variety of systems to be used as controllers such as personal computers, embedded systems, industrial computers or even microcontrollers.

Implementation was developed and designed to work on two platforms – embedded system and personal computer. It is assumed both platforms are using POSIX compatible operating system.

### **Embedded system**

Embedded system is type of computer system designed to be part of larger electronic, robotic or mechanical system. Embedded systems implements various industrial protocols – UART, required for communication with XBeePro-868 module, is usually built-in.

### **RaspberryPi**

The embedded system of choice for the thesis is RaspberryPi [\[Ras15\]](#page-87-11). It was not designed as typical industrial embedded system, but small, extremely affordable computing platform. One piece of RaspberryPi, platform capable of performing around 900MIPS [\[Lon14\]](#page-87-12), costs \$25-\$35 depending on configuration. Most recent specifications of the platform can be obtained at RaspberryPi Foundation website at <http://raspberrypi.org/products/>.

Even if RaspberryPi was designed for consumer and educational market, it still follows principle of embedded system. Various interfaces are available (i.e. UART, GPIO, SPI or Ethernet), however some of them are not available in every model.

Four different models were available at time of the writing. Model  $B+$  (visible on figure [3.4\)](#page-31-0) was chosen as it was widely available at this time, yet any model of RaspberryPi is suitable as control system. Processor architecture of RaspberryPi is ARMv6, so ARMv6 compatible Linux operating system is installed.

### **Connection**

Serial port can be easily accessed on RaspberryPi. Voltage levels both on XBeePro-868 and RaspberryPi are the same 3.3V CMOS level. Therefore it is possible to connect the devices directly without level translator – example pin connection is provided in table [3.5.](#page-31-1) For development purposes connection using jumper cables can be made, however for industrial or any commercial usage proper carrier board should be designed.

Power supply on the board generates 3.3V, although only about 100mA of current is left for external use. It is possible to power the radio from onboard power supply, however the transmission

<span id="page-31-0"></span>

Figure 3.4: RaspberryPi Model B+ [\[Ras15\]](#page-87-11)

<span id="page-31-1"></span>Table 3.5: Connection between RaspberryPi Model B+ and XBeePro-868

RaspberryPi	XBeePro-868		
Function	Pin	Pin	Function
3.3V power supply			VCC
TX signal		3	Data output
RX signal	10	$\mathfrak{D}$	Data input
Ground	39	10	Ground

power should be limited to 1mW. Usage of greater transmission power requires external power supply.

## **Personal computer**

Personal computer is general purpose computer available for individuals. Any modern x86 platform with POSIX operating system (such as a distribution of Linux or commercial OS X system) should be suitable as control system for XBee module. However modern computers do not incorporate RS-232 serial port nor UART CMOS 3.3V level serial port.

Though personal computer is ideal platform for routing application execution, usage of adapters and protocol translators is usually required. USB is standard interface available in almost all personal computers. There are commercially available USB to UART converters and they can be used to connect XBeePro-868 module to the computer. In addition carrier boards specifically made in mind of connection XBee series devices with computers using USB are commercially available.

### **Digi XBee Development Board**

XBee devices manufacturer Digi provides complex developer board (manufacturer part number *XBIB-U-DEV* ) capable of communication with computer using USB (pictured in figure [3.5\)](#page-32-1). Board is supplied with external power source so usage of full transmission power is possible.

The development board can be used to check every feature of XBeePro-868 devices, however consulting its guide [\[Dig11b\]](#page-86-11) is recommended as the thesis does not exploit every feature.

Two pieces of Digi XBee Development Board were used during testing and implementation process of the thesis.

<span id="page-32-1"></span>

Figure 3.5: Digi XBee Development Board[\[Dig11b\]](#page-86-11)

<span id="page-32-2"></span>

Figure 3.6: Sparkfun XBee Explorer Dongle[\[Spa13\]](#page-88-2)

### **Sparkfun XBee Explorer Dongle**

Independent manufacturer Sparkfun produces XBee Explorer Dongle (pictured in figure [3.6\)](#page-32-2)[\[Spa13\]](#page-88-2). It is much simpler device than previously described development board – it connects directly into USB port and drains power from it. Transmission at full power is not recommended, however it possible, depending on USB power capabilities of the personal computer. Main advantage of this device is price, it is 5 times cheaper than competitor's device.

Two pieces of Sparkfun XBee Explorer Dongle were used during testing and implementation process of the thesis.

### <span id="page-32-0"></span>**3.2 Software implementation**

The protocols proposed in the thesis are implemented in  $C++11$  language for POSIX platforms (implementation was tested on Linux  $x86.64/arm11$  and OS X  $x86.64$  machines). C++11 language is chosen for its efficiency, low level system integration and well provided object oriented paradigm. Compared to previous editions of the language, new features such as lambda expressions, native threads or unordered sets simplifies overall implementation. The code is portable and works in heterogeneous environment. Build process and dependencies retrieval is automated with Cmake, subversion and git systems. The most recent version of toolchain is recommended, yet the software was developed using LLVM 3.5 and G++ 4.8 compilers.

*Router* classes and structures are contained in PUT::CS::XbeeRouting namespace. *Router* code is divided into functional sections such as:

- *Driver* provides communication between network and client applications,
- *Device connector* parses or serializes *Frames* and communicates with XBee device over serial port,
- *• Router* routes *Packets* and maintains *Network* topology,
- *• Dispatcher* ensures delivery of *Packets* and keeps them in *History*.

Language of choice for the Simulator was Ruby  $2.2 -$  its great standard library and flexibility allowed to implement portable simulator code in readable manner.

*Simulator* code is divided into  $DSL<sup>2</sup>$  $DSL<sup>2</sup>$  $DSL<sup>2</sup>$  providers, low-level part, data structures and logic.

### <span id="page-33-0"></span>**3.2.1 Building process**

Code developed in the thesis is available online in git repository at [https://github.com/alfanick/](https://github.com/alfanick/xbee868-routing) [xbee868-routing](https://github.com/alfanick/xbee868-routing) or [https://git.amanointeractive.com/poznan-university-of-technology](https://git.amanointeractive.com/poznan-university-of-technology/xbee868-routing-code/tree/master)/ [xbee868-routing-code/tree/master](https://git.amanointeractive.com/poznan-university-of-technology/xbee868-routing-code/tree/master). It can be provided in other form upon request by any of the authors.

### **Dependencies**

Building process is designed to be as simple as possible and requires from user minimal dependencies.

#### **Driver and router**

Essential requirements are:

- POSIX compatible operating system such as Linux  $2.6+$ , FreeBSD 10.0+, OS X 10.5+ low-level system calls are based upon POSIX standards,
- *•* C++11 compiler (clang 3.5+ <http://llvm.org/releases/download.html#3.5> or g++  $4.8+$  – <https://gcc.gnu.org/mirrors.html>) – compiler used to generate binaries for given system,
- *•* Cmake 2.8.8+ <http://cmake.org> portable out-of-source building system,
- Redis 2.6+ server <http://redis.io/download> publish-subscribe pattern provider,
- Git client <http://git-scm.com> distributed version control system used for thesis code and its dependencies,

<span id="page-33-1"></span> ${}^{2}$ DSL – Domain Specific Language

• Subversion client – <https://subversion.apache.org> – version control system used for dependencies retrieval.

Following dependencies are obtained automatically by Cmake:

- *•* Hiredis <https://github.com/redis/hiredis> Redis client library for C/C++,
- google-glog <https://code.google.com/p/google-glog/> Logging library for  $C/C++$ ,
- googletest <https://code.google.com/p/googletest/> C++ unit testing toolchain.
- googlemock <https://code.google.com/p/googlemock/> mocking library used in unit testing.

### **Simulator**

Simulator requires Ruby 2.2+ implementation (standard MRI Ruby [https://www.ruby-lang.](https://www.ruby-lang.org/en/downloads/) [org/en/downloads/](https://www.ruby-lang.org/en/downloads/) is recommended) with croupier-rb 2.0+ ([https://github.com/croupiers/](https://github.com/croupiers/croupier-rb) [croupier-rb](https://github.com/croupiers/croupier-rb)) and redis-rb (<https://github.com/redis/redis-rb>) gems<sup>[3](#page-34-1)</sup>.

### **Building**

After obtaining source code and fulfilling essential requirements run ./configure while being connected to the Internet. The script will generate build files for given platform. Additional dependencies will be downloaded and built.

To build the driver, router and tests use  $./$ build.sh – binaries will be generated inside bin/ directory.

### <span id="page-34-0"></span>**3.2.2 Data structures**

In order to truly represent processed data and simplify algorithms and their implementations assortment of data structures must be introduced.

As the thesis covers low-level processing and data, using types which clearly specify their size is preferred (such as uint8<sub>t</sub>, uint16<sub>t</sub> or uint64<sub>t</sub>. To make code follow real world data types some aliases were introduced as visible in table [3.6.](#page-34-2)

<span id="page-34-2"></span>



### **Frame**

Frame is lowest level data processed in the *Router* software. Frames allow standardized bidirec-tional communication with Xbee device over serial port. XBee module manufacturer<sup>[4](#page-34-3)</sup> specifies complete API manual [\[Dig14b\]](#page-86-4). Frame consists of general API Frame [\[Dig14b,](#page-86-4) p. 35] which encapsulated API-Specific Data (figure [3.7\)](#page-35-0).

<span id="page-34-1"></span><sup>3</sup>'gem' is name for user provided code library in Ruby environment

<span id="page-34-3"></span><sup>4</sup>Digi International, Inc. – <http://www.digi.com>

<span id="page-35-0"></span>

Figure 3.7: API Frame structure[\[Dig14b\]](#page-86-4)

Frame structure is inspired by frame specification written by the manufacturer [\[Dig14b\]](#page-86-4). Therefore struct Frame exists mimicking API Frame (as seen in listing [3.1\)](#page-35-1) and encapsulates union Frame::data consisting of API-Specific Data depending on given enum class Frame::Type. Relationships between type of frames and their functions are presented in table [3.7.](#page-35-2)

Listing 3.1: Simplified Frame structure

```
typedef struct Frame {
  // size of data union
 uint16 t length;
  // identifier of data frame struct (actual type is enum class Type : uint8_t)
 uint8_t type;
  // encapsulated frame struct
 union {...} data;
  // checksum of the frame
  uint8_t checksum;
}
```


<span id="page-35-2"></span>

Implementation of *Router* requires usage only of some of the frames:

- *•* ModemStatusFrame used for discovery device availability,
- *•* CommandFrame used for sending AT commands for configuring and managing XBee device,
- *•* CommandResponseFrame used for AT commands response,
- *•* TransmitFrame used for sending Packets,
- *•* StatusFrame used for acknowledge of transmission status and its statistics,
- *•* ReceiveFrame used for receiving Packets.

Each TransmitFrame and StatusFrame require locally unique identifier (Metadata::frame id) so they can be paired together when status for given transmission is received. Identifier is one byte long, so total of 255 frames could be transmitted before their statuses are received (for identifier value 0 no StatusFrame will be generated).

## **Serialization**

Filled out Frame can be serialized to sequence of bytes using **unsigned char**\* Frame::serialize(**int** & length). Such string can be transmitted and interpreted in XBee device. Serializing method uses memcpy call to serialize Frame::data to sequence of bytes, yet it is important to realize differences in byte ordering. Compiler is let know to omit memory paddings with appropriate #pragma calls. Hence serialization is much quicker than manual conversion using separate bit operations. During serialization, start byte 0x7E is added, length is amended accordingly and checksum is calculated. Resulting string is compliant with XBee documentation.

## **Deserialization**

Deserialization process is parallel to serialization. Sequence of bytes can be deserialized into existing Frame using **void** Frame::unserialize(**unsigned char**\* frame). Length of the string is determined using second and third byte, as it is a part of API Frame specification. Deserialization uses memcpy function to insert bytes into union Frame::data, same cautions about byte ordering are liable. After unserialization it is possible to read specific union field depending on Frame::type and process received data.

## **Packet**

Packet is the only data type exchanged on network level. Packet is encapsulated as data into ReceiveFrame or TransmitFrame when received or transmitted. Packet structure is designed in similar manner as Frame structure (as in listing  $3.2$ ). Every packet can transmit different data according to its type. However the data stored in union data are usually primitive types (as seen in table [3.8\)](#page-37-0).

Listing 3.2: Simplified Packet structure

```
struct Packet {
 // packet type (implemented as enum class Type : uint8_t)
 uint8_t type;
 // source address (address of last visited node)
 Address source;
 // destination address
 Address destination;
 // origin address (address of first visited node)
 Address origin;
```

```
// number of quanta of data if applicable (depends on type)
 uint8_t length;
  // port tag associated with packet
 uint8_t port;
 // data depending on packet type
 union {...} data;
 // mac address of source
 uint64_t mac;
 // constant packet id generated on node where its value is 0
 uint8_t packet_id = 0;
 // status used when packet is Ack (0 means delivery, other value is Address of node where error
      was noticed)
 Address status = 0;
  // list of visited nodes during transmission of the packet
 Path visited;
};
```
#### Table 3.8: Network Packet types

<span id="page-37-0"></span>

Packet can be serialized into Frame encapsulating TransmitFrame using Frame\* Packet::to\_frame (). Serialization is done in different ways depending on  $Frame::type - it$  is achieved using byte operations and previously mentioned memcpy procedure where possible. Deserialization is done in reversed way, comparably to serialization – packet can be deserialized from Frame (usually containing ReceiveFrame) into existing Packet using Packet::Type Packet:from\_frame(Frame\* frame).

Constant uint64\_t Frame::BROADCAST is defined and contains local broadcast address for sending packets adjacent to local device.

## **Packet identifier**

Packet identifier is globally unique in the network until corresponding acknowledgement packet is received. It can be acquired using PacketId Packet::id() function.

Packet identifier is generated from destination, source and packet id fields. Origin node sets packet id to generated Metadata::frame id which is locally unique. Therefore tuple (destination, source, packet id) is unique in whole network and allows to distinguish packet on any intermediate node.

## **Internal packet**

Internal packet is used to encapsulate Frame so they can be analyzed in higher layers such as *Dispatcher* or *Router*. Internal packet is generated when in Router::receive() run across Frame different then ReceiveFrame.

## **Acknowledgement packet**

Acknowledgement packet is sent from destination node to origin node via original packet visited path. On each visited node struct RemoteParameters (listing [3.3](#page-38-0) is appended to the packet. Statistics are used to update antireliability measure on intermediate links.

It is important to notice that Packet::packet id for acknowledgement packet and corresponding data packet are the same, so are the computed PacketId Packet::id() values.

Listing 3.3: RemoteParameters structure

```
struct RemoteParameters {
  // destination address of the statistics
 Address hop = 0;
 // given transmission edge delay in miliseconds
 uint16_t delay = 0;
  // number of undelivered packets during given transmission
 uint8 t errors = 0;
  // number of packet retries on given edge
 uint8_t retries = 0;
};
```
Each of parameters can max out, but never overflow. Knowledge of origin address and each of the hops addresses allows to update antireliability measure on every intermediate link.

Other case when Acknowledgement packet is sent is when transmission error occurs – Packet::status represents address of last node visited. RemoteParameters are still collected, yet they do not reach destination node.

## **History**

History is responsible for saving data packets in memory until their acknowledgement packets are received. If acknowledge is not received before timeout or its status is negative, retransmission of the packet is possible without any actions from client applications. Even if retransmissions fail, data stored in history can be used to return the message back to the sender application.

History keeps relationship between packets Metadata and their Packet::id() and Metadata::frame id (as can be seen in listing [3.4\)](#page-38-1). Metadata store various parameters (listing [3.5](#page-39-0) such as send time, local frame identifier, timeout or retransmission counters. These values are used to control packet routing and maintain safe delivery.

Listing 3.4: Simplified History class

```
class History {
 private:
   // map containing relationship between packet identifier and metadata
   PacketsHistory packets_history;
```

```
// map containing relationship between local frame identifier and metadata
 FramesHistory frames_history;
  // bitset guarding frame identifier occupation
 bitset id_occupation;
public:
  // add packet to history give it identifier
 uint8_t watch(Packet* packet, Path path, time_point timeout);
  // retrieve metadata by packet identifier
 Metadata* meta(Packet* packet);
  // remove metadata from history by packet identifier
 Metadata* erase(Packet* packet);
  // release frame identifier (after status frame is received)
  void release_id(Frame* frame);
```


```
struct Metadata {
  // data packet whose metadata are stored
 Packet* packet;
  // computed packet identifier
  uint32_t packet_id;
  // history of paths (path can be different with every retransmission)
  std::vector<Path> path_history;
  // current Xbee frame identifier
  uint8_t frame_id;
  // aggregated statistics of the packet
  RemoteParameters frame_status;
  // time of last packet transmission
  time_point send_time;
  // calculated timeout for acknowledgement packet
 time_point timeout;
  // number of retransmission
  atomic_ushort retransmission_counter;
\vert } ;
```
*Dispatcher* uses History::watch() to compute next free frame identifier and add the packet to the History. It also used to store and check acknowledge timeout and retransmission counter. When Packet is delivered or timeout expires *Dispatcher* removes Packets and its Metadata from History.

## **Node**

<span id="page-39-1"></span>Node is representation of single network node (listing [3.6\)](#page-39-1).

Listing 3.6: Simplified Node class

};

```
class Node {
  public:
    // network address
    Address address;
    // mac address (applicable only for adjacent nodes)
    uint64_t mac;
    // node name (applicable only for self node)
    string name;
    // marks self node
    bool self;
};
```
Every vertex inside Network graph contains Node. Self node is special case and only one self node can exist in the Network. Only self string Node::name is known to router, as it requires AT command to obtain the name, however uint64\_t Node::mac addresses of direct neighbours are known and Address Node:: Address are known for every discovered Node in the Network.

#### **Network**

Network represents graph containing connections between nodes. Every vertex incorporate Node and for every edge associated is Parameters structure (listing [3.7\)](#page-40-0).

Network is represented as undirected graph. Internally graph is represented as unordered\_map< Address, unordered\_map<Address, Parameters\*>> as compromise between access complexity and memory usage.

Listing 3.7: Edge Parameters structure

```
struct Parameters {
 // total number of delivered packets
 uint32_t good = 0;
 // total number of undelivered packets
 uint32_t errors = 0;
 // total number of packet retries
 uint32_t retries = 0;
 // averaged delay on given edge (in milliseconds)
 uint16_t delay = 10;
 // antireliability measure based on prior parameters
 float antireliability();
};
```
<span id="page-40-1"></span>Network constructor requires self node address, so first node can be created and accessed using Node\* Network::self(). Elementary operations such as adding or removing nodes or edges are implemented (listing [3.8\)](#page-40-1). Operation of merging two graphs is implemented, as it is used with Packet::Type::NodeBroadcast packet. Overflow-safe updating of edge Parameters is implemented using Network::update(). Summarized list of operations available in Network is presented in listing [3.8.](#page-40-1)

Listing 3.8: Simplified Network class

```
class Network {
  public:
    // constructor - adds self node
    Network(Address self);
    // try to add new nodes and edges to current graph
    bool merge(Edge* edges, uint8_t length);
    // try to add new edge
    bool add_edge(Address a, Address b);
    // try to add new node
    bool add_node(Address a);
    // check for edge existence
    bool adjacent(Address a, Address b);
    // remove edge
    bool drop(Address a, Address b);
    // get edge parameters
    Parameters* edge(Address a, Address b);
    // get node parameters
    Node* node(Address a);
    // get self node
    Node* self();
    // convert graph to list of edges
    Edge* graph(uint8_t &length);
    // get mac address of given node address
   uint64_t mac(Address a);
    // get node address from mac address
    Address from_mac(uint64_t mac);
    // update edge parameters
    void update(Address a, Address b, uint8_t retries, uint8_t error, uint16_t delay);
    // find most reliable path between two nodes without visiting already visited nodes
    Path path(Address from, Address to, Path visited);
};
```
## **Essential operations**

Essential operations such as **bool** Network::add\_edge(Address a, Address b);, **bool** Network::add\_node (Address a);, **bool** Network::merge(Edge\* edges, uint8\_t length); and **bool** Network::drop(Address a, Address b); do not generate error when node or edge does not exists (when removing) or does exist (when adding). Instead, boolean value true is returned when operation is successful (that is when graph changes after the operation) and false is returned when operation does not change the graph.

Furthermore **bool** Network::add\_edge(Address a, Address b); and **bool** Network::merge(Edge\* edges, uint8\_t length); operations will add missing nodes, so edge between could be created.

#### **Edge parameters update**

Figure 3.8: Delay calculating method

$$
delay = \lceil \frac{delay_{previous} + delay_{new}}{2} \rceil \tag{3.1}
$$

<span id="page-42-0"></span>Edge Parameters are used to calculate antireliability measure and timeouts. Most recent link statistics are preferred. Method **void** Network::update(Address a, Address b, uint8\_t retries, uint8\_t error, uint16\_t delay); is preferred way of updating edge parameters, as it ensures no overflow will happen.

Additionally uint16<sub>\_t</sub> Parameters::delay is calculated as moving average (equation [3.1\)](#page-42-0) – previously calculated delay and current delay (with delay assumed to be 10*ms* at the beginning) are averaged together.

#### **Most reliable path**

As described in the routing algorithm, path between *source* and *destination* is assumed to be most reliable when sum of antireliability measures on intermediate edges is minimal. Therefore to obtain most reliable Path, antireliability measure is treated as edges weight and path with smallest cost is pretended to be the most reliable one. To avoid cycles previously visited nodes should not be visited again. List of previously visited nodes is included when computing path and cost of visiting already visited nodes is increased drastically.

To compute such Path well known Dijkstra shortest path algorithm [\[Dij59\]](#page-86-0) is used. Implemen-tation based on priority queues [\[Bar98\]](#page-86-1) allows to efficiently solve shortest path algorithm between two nodes in the network. Built in std:: priority\_queue is used with custom comparator based on Parameters::antireliability().

# <span id="page-42-1"></span>**3.2.3 Driver**

*Driver* provides way for any application to communicate with the network based upon routing system implemented in the thesis. It is a layer of abstraction hiding hardware, routing operations and network topology.

Well designed driver ensures following concepts:

- adds minimal performance overhead,
- *•* allows bidirectional communication with any device in the network,
- hides network topology,
- has minimal dependencies,
- *•* works in heterogeneous environment,
- easily integrates with client applications.

## **Publish-Subscribe pattern**

Publish-Subscribe pattern allows publisher to send message to given channel for any number of clients, furthermore any number of clients can listen for any channel, even if there is no publishers in observed channel [\[CMPC04\]](#page-86-2).

To be specific Topic-Based Publish-Subscribe pattern [\[EFGK03\]](#page-86-3) is used in *Driver* implementation as inter-process communication tool. In this approach publisher labels every message with topic, while can subscribe to patterns which can match any number of topics.

Following topic naming convention is used internally by *Driver*:

# network:PORT:DESTINATION:SOURCE undelivered:PORT:DESTINATION:SOURCE

Where PORT is number in range [1*,* 255], DESTINATION and SOURCE are accordingly destination and source addresses. Addresses can be numbers in range [1*,* 255] or token self which explicitly marks local node.

Any of the parameters can be replaced with \* (asterisk) – the matching operator. Therefore it is possible to subscribe and receive message with any combination of tuple (PORT, DESTINATION, SOURCE). Interesting and useful patterns with explanations are provided in table [3.9.](#page-43-0)

<span id="page-43-0"></span>

#### Table 3.9: Example patterns

#### **Redis**

Redis is open source in-memory key-value database and distributed implementation of Publish-Subscribe pattern. It has low memory footprint and low performance overhead, although it does implement Publish-Subscribe with topic pattern matching.

Using  $C/C++$  driver Hiredis and few commands (as shown in table [3.10\)](#page-44-0) satisfies requirements for *Router*. For each machine only one instance of Redis server is required, even if there is more than one *Router* instance (as happens on developer machines or when using *Simulator*). To prevent topic collisions, every *Router* may introduce each own prefix (a number in range [1*,* 255], generating ie. topic 13/network:\*:\*:self instead of network:\*:\*:self).

Usage of Redis comes up with additional property – it is possible to connect *Driver* to remote instance of Redis server. As a consequence it is feasible to connect multiple remote applications to local machine with *Router* and XBee device. Multiple complex topologies can be created with this feature as shown by examples in figure [3.9.](#page-44-1)

<span id="page-44-0"></span>



<span id="page-44-1"></span>

(a) Multiple applications connected to multiple routers using Ethernet



(b) Possible sport timing system architecture

Figure 3.9: Example networking topologies

## **Implementation details**

*Driver* part is implemented with three classes:

- Driver bridge between C++ and network, most important part from developer point of view,
- *•* Manager creates Subscriptions with given topic prefix,
- *•* Subscription subscribes to single Redis topic, runs in separate thread.

Driver interacts with applications in asynchronous matter. C++11 language features such as std::function (extension of function pointer concept) is used to bind code with message receipt events – any type of code such as functions, methods or lambda can be used.

Driver defines function types as follows:

## Listing 3.9: Driver handlers types

```
// Exhaustive handler with destination address, port, source address, data and data length
typedef std::function<void (Address, uint8_t, Address, uint8_t*, size_t)> Action;
// Simplified handler with source address, data and data length
typedef std::function<void (Address, uint8_t*, size_t)> LocalAction;
```
Constant Driver::SELF is defined and recognized in Driver. It should be used to mark or check local address.

With previously introduced handler types following Driver methods connect network with code:

- $\bullet$  void self(Action action); executes action when any message to self is received,
- *•* **void** listen(uint8\_t port, LocalAction action); executes action when message on given port to self is received,
- *•* **void** deliver(Address destination, uint8\_t port, uint8\_t\* data, uint8\_t length); sends given data to destination on given port,
- **void undelivered(Action action);** executes action when message from self could not be delivered.

These methods abstract network topology and hide routing, yet they follow the concept of reliable network implemented in the thesis. It is possible to *send* messages, *receive* messages and get notified of *undelivered* messages.

Presented code listing [3.10](#page-45-0) shows different ways to interact with the network in echo style application. It listens on port 13 and sends back to source any received message, in case of delivery failure handle error is executed. Overall three threads are created – one for main function, one for listener and one for undelivered messages listener. Program is running until killed from outside environment.

Listing 3.10: Example echo application

```
#require <iostream>
#require <unistd.h>
#require <driver/driver.h>
using namespace std;
using namespace PUT::CS::XbeeRouting;
// undelivered messages handler
void handle_errors(Address destination, uint8_t port, Address source, uint8_t* data, size_t length)
     {
  cerr << "Could<sub>u</sub>not<sub>u</sub>deliver<sub>u</sub>data:<sub>u</sub>'" << data << "'<sub>u</sub>to<sub>u</sub>" << destination << endl;
\vert }
// basic echo program
int main() {
  // driver instance
  Driver driver;
  // listen for incoming messages on port 13 (lambda function)
  driver->listen(13, [](Address source, uint8_t* data, size_t length) {
    // send back data
    driver->deliver(source, 13, data, length);
    cout \ll "Received<sub>\cup</sub>," \ll data \ll "'\simfrom\sim" \ll source \ll endl;
  });
  // handle undelivered messages (function pointer)
  driver->undelivered(handle_errors);
  // let other threads execute
```
pause(); **return** 0; }

*Driver* is build as shared library available in bin/libxbee network.so (or libxbee network.dylib on OS X platforms). By default Driver connects to local Redis server instance by Unix socket at path /tmp/redis.sock. *Driver* can be further configured using environment variables presented in table [3.11.](#page-46-0) Any program using *Driver* shared library reacts to previously mentioned environment variables.

Table 3.11: *Driver* environment variables

<span id="page-46-0"></span>

Variable	Type	Default	Description
REDIS HOST	network address	none	network address of Redis server (using)
			Unix socket if not provided)
REDIS_PORT	number	6379	Redis server port
REDIS_DATABASE	number	none	prefix of topics in case of using multiple
			<i>Router</i> instances with same Redis server

## **3.2.4 Dispatcher**

Dispatcher is responsible for successful delivery of Packet or discovering delivery failure. Instance of Dispatcher is created by Router. Implementation of Dispatcher realize diversity of operations as seen in listing [3.11.](#page-46-1)

Listing 3.11: Basic Dispatcher class

```
class Dispatcher {
 public:
   // constructs instance of dispatcher with access to XBee device, network graph and driver
   Dispatcher(Xbee& x, Network& n, Driver& d);
   // scan incoming packet
   void scan(Packet* p);
   // analysis of data packet
   void handle_data(Packet* p);
   // analysis of acknowledge packet
   void handle_ack(Packet* p);
   // analysis of internal packet (other than receive xbee frame)
   void handle_internal(Packet* p);
   // send packet with delivery guarantee
   bool deliver(Packet* p);
   // send packet to adjacent node
   bool send(Packet* p);
   // locally broadcast packet
   void broadcast(Packet* p);
   // check for timeouted packet
   int tick();
   // retransmit packet
```

```
bool retransmit(Metadata* meta);
    // retransmit packet if number of retransmissions is not maxed out
    bool try_retransmit(Metadata* meta);
};
```
Dispatcher can scan every incoming Packet for data required to guarantee successful packets delivery. Moreover it is charge of updating network parameters, based on statistics carried within acknowledge packets.

#### **Packet analysis**

Every Packet received by Router is scanned by Dispatcher::scan() before further processing. Various parameters are obtained depending on packet type. To make implementation more readable, handling method depending on packet type were introduced.

#### **Data packets**

Packet is considered as successfully delivered when Packet destination is self node. In this case positive acknowledge packet is generated and sent towards source node.

# **Internal packets**

Internal packets encapsulated XBee Frame other than ReceiveFrame. Frame of type StatusFrame is processed inside Dispatcher::handle\_internal(Packet\* packet) as it contains delivery status information of last send Frame.

Every StatusFrame is associated with frame id, that being so Metadata is retrieved from the History. Statistics of given Packet, stored inside metadata, such as delay, number of errors or retries are updated. Using the same statistics parameters of link between self and status source are updated.

If delivery status is negative, packet retransmission occurs (using Dispatcher::try retransmit), unless retransmissions are no allowed any longer for this packet. In this case, negative acknowledge packet is sent towards packet source.

Receipt of positive acknowledge packet causes corresponding Packet and its Metadata to be removed from History, as they no longer be useful.

#### **Acknowledge packets**

Acknowledge packets have the same PacketId as corresponding data packet, so Metadata is easily retrieved from the History.

Acknowledge packet carries status and edge parameters in form of array of RemoteParameters. Local Network graph is updated with these statistics. Every acknowledge packet passing through intermediate node refreshes local graph, so statistics used to calculate antireliability measures are most recent.

Local node parameters retrieved from History for associated packet are appended to acknowledge packet. If local node is not associated packet source (origin of acknowledge packet) it is passed to next node (using Dispatcher::send()) following the same path the original packet was routed.

Otherwise if acknowledge packet reached its destination and its status is negative, retransmission of corresponding data packet occurs (if it is still allowed).

## **Transmission routines**

Different packet transmission routines are implemented (as seen in table [3.12\)](#page-48-0). Method Dispatcher::deliver() is most important in context of the thesis, as it ensures Packet delivery.

<span id="page-48-0"></span>

Method	Range	Description
send()	physical neighbours	Send packet to locally adjacent neighbour without any
		guarantee of success
broadcast()	physical neighbours	Send packet to every locally adjacent neighbour (broad-
		cast) without any guarantee of success
deliver()	any node in network	Send packet to any node in the network, guarantees de-
		livery or information in case failed delivery

Table 3.12: Dispatcher transmission methods

Both send and broadcast methods basically encapsulate Packet in TransmitFrame with proper MAC address and do not receive StatusFrame in response. Message is delivered back to Redis using Driver::deliver back() in case the source node is not adjacent to destination node (message can be recognized as undelivered in client applications).

## **Delivery commitment**

Delivery commitment in deliver, implemented accordingly to algorithm described in previous chapter, requires more processing. Most reliable path is calculated using network.path(self->address , packet->destination, packet->visited). If such path could not be found and self node is source node (not intermediate one), message is delivered back in same manner as mentioned previously.

Packet is added to History and frame id is obtained using history.watch(packet, path, timeout (packet, path)). Therefore Metadata is created for the packet and its delivery status is monitored. Packet's acknowledgement must be received within timeout otherwise, after number of consequent failures packet will be considered as lost. It will be delivered back to Redis as undelivered if failure is detected on source node, otherwise acknowledge packet marking failure will be send back to source, causing delivering back to Redis.

#### **Packet retransmissions**

Method **bool** Dispatcher::retransmit(Metadata\* meta) was designed to support Packet retransmission. It is implemented comparably to deliver method, however as the Packet already exists in the History there is no need to use History::watch(). Instead new Metadata::frame\_id is issued and updated in the History. Both new Metadata::send\_time and Metadata::timeout are assigned. Packet is serialized and encapsulated into Frame, as such it is sent to XBee device.

Retransmission count of given packet is restricted by constant Metadata::retransmission\_max = 5. Helper method **bool** Dispatcher::try\_retransmit(Metadata\* meta) was created which retransmit Packet stored in the History. With each retransmission adequate Metadata::retransmission counter is incremented, until retransmission count reach retransmission max. In such case the Packet is assumed to be undelivered and it is returned to client application.

#### **Timeout system**

Constructor of Dispatcher class creates background thread responsible for detecting data Packet with expired acknowledge timeout.

In this thread **int** Dispatcher::tick() method is executed periodically (with period set as miliseconds Dispatcher::tick\_sleep\_time = 100). Timeout is calculated for every Packet individually using

time\_point Dispatcher::timeout(Packet\* p, Path& path) as part of deliver method. Timeout method calculates deadline for acknowledge packet to return, accordingly to equation [2.2](#page-21-0) – timeout depends upon Path to destination and intermediate link statistics such as delay, retries, errors and deliveries count.

For each Packet contained in History previously computed timeout is compared with current timestamp. If acknowledge packet is not received before timeout, retransmission is performed unless maximal retransmission tries was conducted. In this case the packet is marked as undelivered and its Metadata is removed from History.

## **3.2.5 Router**

Router class is entry point of router application. It is responsible for routing process and topology discovery. Router creates connection with physical XBee device and obtains basic parameters from the device. Router works in simple processing loop as shown in listing [3.12.](#page-49-0)

Listing 3.12: Router processing loop pseudocode

```
while (should_run) {
  // receive next frame from xbee
  frame = xbee.receive();
  // deserialize packet
  packet = packet->from_frame(packet);
  // allow dispatcher to process the packet
  dispatcher.scan(packet);
  // data packet
  if (packet->type == Packet::Type::Data) {
    // if packet reached its destination
    if (packet->destination == self) {
      // publish it to redis so other applications can read the message
      driver.publish(packet);
    }
    // if this is intermediate node
    else {
      // let dispatcher move packet forward
      dispatcher.deliver(packet);
    }
  }
  // topology packet
  else {
    // analyze packet message so network topology can be updated
    proceed_with_topology_maintenance(packet);
  }
}
```
The pseudocode shows basis of operation, however, Router code is more complex (listing [3.13\)](#page-49-1) and closely reflects algorithms presented in previous chapter.

Listing 3.13: Basic Router class

```
class Router {
 // constructs router with xbee device connected to given serial port and given address as self
      node address
 Router(char* serial_port, Address address);
 // creates router instance and processing loop
```

```
static void run(char* serial_port, Address address);
  // receive and deserialize packet from xbee
 Packet* receive();
 // process next packet, behaves adequately to packet type
 void process();
  // broadcast heartbeat packet to let other nodes know about its existence
 void heartbeat();
};
```
## **Initialization**

Instance of XBee device connector (described in previous section) is created. Radio power level is configured and basic parameters such as node name, network address and local MAC address are obtained. As a result Node\* Network::self() is configured.

Router listens on channel network: \*: \*: self using Driver:: self, so messages that need to be sent to other nodes from external applications can be intercepted and delivered using Dispatcher::deliver().

Additional thread, broadcasting Packet::Type::NodeBroadcast every 15 seconds acts as heartbeat process and is required for proper topology discovery and maintenance process.

#### **Receiving process**

Packet is received from XBee device using blocking method Packet\* Router::receive(). Frame acquired from XBee device is analyzed. Packet is deserialized from Frame::Type::Receive, however, other types of Frame cause Packet::Type::Internal to be generated and surround the frame.

## **Processing packets**

After receiving the packet, it is let to be processed by Dispatcher in the first place. Then, packet is processed adequately to its type. Some packets carry data, other one can be acknowledge or topology maintenance packets. switch statement is to differentiate actions for the packet.

#### **Data packet**

Data packet can be processed in two ways, depending on its destination. If given node is packet destination (that is self->address == packet->destination), then contained data is published to Redis using Driver::deliver method. At this moment acknowledge packet has been already sent by Dispatcher.

Otherwise, given node is intermediate node on path to packet destination. As so, packet must be routed to its destination. After adding Address self->address to list of packet waypoints, the packet is sent further using Dispatcher::deliver().

## **Acknowledge packet**

On Router layer, acknowledge packets are used just for logging purposes.

#### **NodeBroadcast packet**

NodeBroadcast packets are received as Node heartbeat. As designed in topology discovery algorithm, edge between self node and node address contained in the packet is added to the Network graph. If graph was updated (that is self node did not have the edge in the graph already), current state of the graph is locally broadcasted.

### **EdgeDrop packet**

EdgeDrop packets cause removal of Edge contained in the packet from self Network graph. Packet is further propagated (using Dispatcher::broadcast), only if the connection defined Edge was existing in the graph.

#### **Graph packet**

Graph packets are received from adjacent nodes after successful recognizing of new node. It is assumed that received graph may contain new nodes and edges. As so Network::merge operation is executed. If state of the local Network graph was changed, updated local graph is broadcasted.

## <span id="page-51-0"></span>**3.2.6 Simulator**

In order to make design and development process more reliable XBeePro-868 simulator was developed. Algorithm and implementation should be tested on various network topologies and in different environments. Such tests are not possible with five radio modules owned by authors. Tests could have been conducted using larger amount of XBee modules, however it is not an economical solution. In addition these tests could not have been reliable as real world environment can behave in unpredictable way.

Therefore simulator was created – one instance of the simulator can simulate network of physical XBeePro-868 nodes. The simulator uses POSIX feature such as pseudo terminals (called PTY). On POSIX platforms physical serial ports are represented as special files available as /dev/tty\* path. Similarly PTY are special files (usually available as /dev/pty\*) with similar behaviour to serial port – read and write operations are available on both sides of the PTY. As a result it is possible to deceive the router application and pass path to pseudo terminal instead path to real serial port – no changes in router code are required.

Simulator is developed in Ruby language, because of its great standard library and vast available community provided code as well as metaprogramming availability. It is possible to use simulator as standalone application using bin/simulator or as library used in other Ruby code. When used as standalone application list of paths to PTY and names of associated virtual XBeePro-868 devices is printed to standard output, various logs are printed to standard error stream.

Simulator splits network topology from environment definition. Network topology is definition of undirected graph with possible links between XBeePro-868 devices. Environment is file describing behaviour of nodes and links at any moment of time. Parameters such as node power state, link delay and number of frame retries and errors can be configured.

#### **Basis of operation**

In order to use simulator, network topology must be provided. Network topology is represented by undirected graph, where vertices represent XBeePro-868 module (identified by name) and edges represent aerial links between the modules. There is no theoretical limit on number of nodes or edges, however simulation efficiency can decrease when large dense graph is created

<span id="page-52-0"></span>

Parameter	Subject	Value	Description
Power	node	boolean	Power state of the node. Powered off devices cannot receive
			or transmit frames.
Retries	edge	number	Number of frame retries required to transfer frame between
			two nodes.
Delay	edge	number	Time necessary to transfer one frame between two nodes (for
			each retry delay is accumulated).
Errors	edge	boolean	Determines if frame is delivered when using given link.

Table 3.13: Environment parameters

Inspired by real world, environment definition must be supplied – environment defines behaviour of nodes and links. In real world node can be either powered on or off, links introduce delay depending on propagation time and distance and various events can contribute to number of transfer retries and errors (parameters summary is available in table [3.13\)](#page-52-0).

Furthermore any of the parameters can be common for every node and link or can be defined for specific node or link. The environment uses directed version of network topology – it is possible to specify different parameters for two directions of same aerial link.

Moreover parameters values can be represented using probability density functions. This approach allows accurate modeling of real world behaviour. Various probability distributions are available, however, Gaussian (normal), Poission and degenerate distributions are most useful in this case [\[Tij12\]](#page-88-0).

Output of probability density functions can be scaled and biased so parameters can be retained in specific ranges. Additionally probability density functions can be specified differently depending on time increasing reality of the simulation's model. Therefore value of specific parameter *p* is defined as value of stochastic process  $X_{p,id}(t)$  in given time point *t* for link or node *id* as presented in following equation:

$$
X_{p,id}(t) = a_{p,id} \cdot f_{p,id}(x,t) + b_{p,id}
$$

Where:

*ap,id* is scale of parameter *p*,

 $b_{p,id}$  is bias of parameter *p*,

 $f_{p,id}(x,t)$  is random value generated using probability density function at given time *t*.

Parameters values are calculated in different ways as they need to be converted to appropriate types using following equations:

• power must be converted to boolean value

$$
power_a(t) = \begin{cases} 0 & X_{power,a}(t) < 1\\ 1 & X_{power,a}(t) \ge 1 \end{cases}
$$

*•* retries cannot exceed maximum number of retries

$$
retries_{a \rightarrow b}(t) = \begin{cases} X_{retries,a \rightarrow b}(t) & X_{retries,a \rightarrow b}(t) < retries_{max} \\ retries_{max} & X_{retries,a \rightarrow b}(t) \ge retries_{max} \end{cases}
$$

*•* delay must incorporate number of retries

$$
delay_{a\rightarrow b}(t) = X_{delay,a\rightarrow b}(t) \cdot (retries_{a\rightarrow b}(t) + 1)
$$

*•* link error parameter must be converted to boolean value

$$
errors_{a\rightarrow b}(t)=\begin{cases}0 & X_{errors,a\rightarrow b}(t)\leq 0\\ 1 & X_{errors,a\rightarrow b}(t)>0\end{cases}
$$

Where:

*a, b* are node identifiers,

*t* is time point,

*retriesmax* is maximum number of frame retries.

It can be noticed that in given time *t* effective networking graph can change. If a node is marked as powered off, depending on  $power_a(t)$ , the node is effectively removed from the graph, so any transmission to the node will fail. Similarly if an edge's  $errors_{a\rightarrow b}(t)$  parameter denotes link failure, transmission from *a* to *b* will fail. Therefore environment configuration can dynamically change effective network graph (examples can be seen in figure  $3.10$ ).

#### **Processing loop**

With presented parameters definition, following processing loop is executed for each node.

A frame is received from given node pseudo terminal. Frame is deserialized from raw byte sequence.

If frame is CommandFrame type appropriate radio parameters are set or read and CommandResponseFrame is sent back to the node.

If frame is TransmitFrame decapsulated data are passed to destination nodes' receive method. In case of broadcast transmission, the data are passed to every adjacent node in effective graph. If transmission can occur, that is effective graph computed using environment state in given time contains edge connecting source and destination node, ReceiveFrame is created.

The data are encapsulated into the ReceiveFrame, after delay depending on  $delay_{a\rightarrow b}(t)$  the receive frame is serialized and sent over destination's pseudo terminal to destination's application. StatusFrame is generated and sent to source node pseudo terminal. In case of transmission failure (depending on  $power_b(t)$  and  $errors_{a\rightarrow b}(t)$ ) the status frame contains appropriate statistics.

Every event in processing loop creates log entry tagged with node name, frame data and time relative to start time of simulation. Probability related functions are implemented using community developed courier-rb library.

## **Input files**

Simulator requires two input files. These files describe network topology and environment. It can be noticed that environment definition is quite detailed and may require complex input file.

In order to make input files human readable YAML file format was chosen. YAML is human readable, indentation based, popular data format. It is assumed syntax in examples provided below is self-explanatory. Complete description can be found in reference card [\[YAM06\]](#page-88-1).

Nodes are identified by name, so any unique number or string can be node identifier.

#### **Network topology**

Network topology is represented as adjacency map – every node is associated with list of adjacent nodes. Example chain topology network can be seen in listing [3.14.](#page-54-1) By convention network definitions are stored in \*.network.yml files.

<span id="page-54-0"></span>

Figure 3.10: Network graph in various moments of time  $% \left\langle \cdot ,\cdot \right\rangle _{0}$  Effective graph is shown in black.

<span id="page-54-1"></span>



As network topology graph is undirected, both direction entries are redundant and may be omitted (as can be seen in listing [3.15\)](#page-54-2).



<span id="page-54-2"></span>alfa: - beta beta:

```
- gamma
gamma:
  - alfa
```
Two different syntaxes can be observed in listings [3.14](#page-54-1) and [3.15,](#page-54-2) presenting two admissible ways of enumerating lists in YAML.

#### **Environment definition**

Environment definition requires file structure which allows to specify common or specific parameters distribution for nodes and edges in given time points. Therefore environment file structures is divided into time range which can start in some absolute time point or after delay relative to previous time point. Definition of example "perfect" environment can be seen in listing [3.16](#page-55-0) – such environment guarantees transfer of every frame.

<span id="page-55-0"></span>

start: point: 0			
edges:			
all:			
delay: 5			
retries: 0			
errors: 0			
nodes:			
all:			
power: 1			

Listing 3.16: Example perfect environment definition

Each time range is labeled with identifier and its start time point must be specified. To specify absolute time moment point key must be used, in order to specify time relative to previous time point delay key can be used.

Most recent parameters definitions are used – if parameter for given node or frame is not defined in current time frame, last definition is used. In each time frame parameters for edges can be specified (in section edges) so as nodes parameters (in section nodes). A key all can be used to set parameters of every edge or node. To set parameters for specific node, its name must be used as key in nodes section. In similar fashion setting link between node a and node b parameter requires to use array [a, b] as key in nodes section.

Parameter distribution can be specified in two ways, depending on required distribution. Degenerate distribution can use simplified syntax as it can be defined using one constant. Other distributions require full distribution definition. To pair parameter with distribution a parameter name must be used as a key and list of properties must be passed, where distribution key is required (for degenerate distribution single number can be passed instead of properties).

Every distribution can use bias and scale parameter, however, any distribution requires other specific constants. Distribution names and their required arguments are dependent on croupier-rb implementation – list of useful distribution names and their parameters can be seen in table [3.14.](#page-56-0) More distribution specifications are available in croupier-rb gem documentation.

Complex example presenting described parameters is available in listing [3.17.](#page-56-1) Three time ranges will occur – start at  $t = 0$ , storm at  $t = 10$  and back to normal at  $t = 60$ . Short syntax for degenerate (constant) distribution, parameters specific for edge [alfa, gamma] and node beta and usage of YAML references can be noticed.

<span id="page-56-0"></span>

Distribution name	Default parameters	Description
normal	$mean = 0.0, std = 1.0$	(Gaussian) distribution with Normal
		given mean and standard deviation.
poission	$lambda = 50$	Poisson distribution with given lambda.
degenerate	$constant = 42$	Discrete distribution which returns the
		same value each time.
uniform	$included = 0.0, excluded = 1.0$	Uniform distribution generating number
		between <i>included</i> and <i>excluded</i> , where
		each value is equally likely.

Table 3.14: Available probability distributions

<span id="page-56-1"></span>

start:		
point: 0		
$\mathtt{edges}$ :		
all: &perfect_edges		
delay: 10		
retries: 0		
errors: 0		
[alfa, gamma]:		
retries:		
distribution: uniform		
included: 0		
excluded: 5		
nodes: &nodes_on		
all:		
power: 1		
storm:		
delay: 10		
edges:		
all:		
delay:		
distribution: uniform		
included: 20		
excluded: 50		
nodes:		
beta:		
distribution: normal		
mean: $-1$		
bias: $-0.2$		
back_to_normal:		
point: 60		
edges:		
<<: perfect_edges		
nodes:		
<<: nodes_on		

Listing 3.17: Complex environment example

#### **Implementation summary**

Simulator code is divided into multiple parts however following classes has crucial application:

- XBee::Node single XBeePro-868 node, communicates over its own pseudo terminal, deserializes frames and responses to various events,
- XBee::Network manages graph of nodes, it is responsible for calculating effective graph and properties,
- *•* XBee::Frame representation of XBeePro-868 frame.

Classes' implementation were divided into multiple partial classes files grouped by implemented functionality type.

#### **Frame implementation**

Implementation of XBeePro-868 frames is required to operate on data serialized as raw bytes sequences. Every frame documented in XBeePro-868 manual [\[Dig14b\]](#page-86-4) is implemented as Ruby class inside XBee::Frame module.

## **Domain Specific Language**

Domain Specific Language was introduced to overcome problem of frames implementation. Domain Specific Languages are computer languages designed to be specialized in distinct problem [\[VDKV00\]](#page-88-2). Human friendly DSL to solve the problem was designed as superset of Ruby language. It is executable Ruby code even though it looks like new programming language.

Listing 3.18: TransmitFrame implementation

<span id="page-57-0"></span>

frame 0x10, :transmit,
id: $'C'$ ,
mac: $Q>2$ ,
network: 'S>',
command: $'a2$ ,
status: $'C'$ ,
data: '*a'

Using various metaprogramming methods, procedure XBee::Frame::frame(id, name, \*\*args) dynamically create class which represents XBee frame identified by id and name. Field structure is specified using \*\*args dictionary – every field is specified by name and  $Array:pack$  compatible string [\[Rub13\]](#page-87-1). Therefore each frame can be specified as in similar way as example TransmitFrame listing [3.18.](#page-57-0) Each frame implementation supports serialization using to bytes and deserialization using from bytes.

#### **Supported events**

Only subset of operations supported by XBeePro-868 device is implemented – the subset covers every frame required for proper router simulation. Implementation reacts to frame received frames such as TransmitFrame and CommandFrame.

Frame are handled inside Node class. Using metaprogramming another DSL feature was provided – XBee::Node::receive(type, \*\*args, &block) and XBee::Node::reply(type, \*\*args) methods.

Using receive method allows to filter incoming frame by its type and by contained data values. It allows to write event-based code as it can be seen in listing [3.19](#page-58-0) – the example presents implementation of handling broadcast-type TransmitFrame.

<span id="page-58-2"></span>





```
receive :transmit, mac: 0xFFFF do |frame|
 broadcast frame.id, frame.data
end
```
Method reply enable creating, serialization and sending frame over pseudo terminal to client application. Example usage showing CommandResponseFrame reply with response to NI command is shown in listing [3.20.](#page-58-1)

Listing 3.20: NI command response

```
reply :command_response,
 id: frame.id,
 command: 'NI',
 status: 0,
 data: @id.to_s
```
## **Supported commands**

Only subset of configuration commands is supported in the simulator. These commands are sufficient set to make router application fully operational. Implemented commands list is available in table [3.15.](#page-58-2) Further commands implementation can be easily added if application requires them.

# **3.3 Testing process**

In order to ensure that the implementation of the network protocols was properly designed various tests has been conducted. Two testing strategies were chosen – unit testing and behaviour testing. Tests has been written concurrently with the implementation, so any flaws has been fixed as soon as they have been found.

# **3.3.1 Unit testing**

Unit testing is concept of testing small individual units of software's source code [\[PM14\]](#page-87-2). Individual methods of implementation are tested in simplified environment. Therefore implementation errors can be found independently and their localizations can be easily determined.

Unit tests has been written with help of googletest library [\[Goo15\]](#page-87-3). *Network* and *Dispatcher* were seriously tested as core parts of the implementation. These tests were partitioned into tests cases (table [3.16\)](#page-59-0) comprehensively testing various combinations of input data. In the end 15 test

Table 3.16: Test cases

<span id="page-59-0"></span>

Name	Testing Goal
timeout	Calculation of proper timeouts depending on collected edge
	statistics.
add_node	Ability to add new node unless node is already in the graph.
node	Accessing node by its address.
nodeForbiddenAddress	Awareness of invalid addresses when adding node or edge.
add_edgeWhenNodesDoNotExist	Creating nodes associated with new edge when these nodes
	are not already in the graph.
add_edgeWhenNodesExist	Addition of new edge to the graph.
add_edgeSelfLoop	Inability to add loopback connection.
dropEdge	Removing edge form the graph.
mergeGraphs	Merging two graphs should result in addition of non existing
	nodes and edges to target graph.
mergeTheSameGraphs	Merging edges already existing in target graph should not
	change the graph.
pathSingleChain	Most reliable path between two nodes should be calculated
	in basic chain graph.
pathNotChainGraph	Most reliable path between two nodes should be calculated
	in complex graph.
pathInTreeGraph	Most reliable path between two nodes should be calculated
	in tree.
pathNoPath	Empty path should be expected when route between two
	nodes cannot be calculated.
pathNoChainGraphVisited	Calculated path should not contain already visited nodes.

cases were formulated with over 120 assertions. Multiple flaws were found and fixed using this testing method.

## **3.3.2 Behaviour testing**

Behaviour testing has been conducted as part of behaviour-driven development process. In this technique various behaviours are described as *scenarios* which consists of preconditions, actions and expected results [\[PM14\]](#page-87-2). The whole system, in our case implementation of network protocols, is tested at once so complex behaviour can be described and examined.

In case of the project, behaviour of protocols' implementation can be studied in various topologies and environment using previously designed simulator. As the simulator was implemented in Ruby language, *Cucumber* application [\[Asl15\]](#page-86-5) has been chosen as it easily integrates with the simulator.

Scenarios of the tests were written in *Gherkin* language [\[Ye13\]](#page-88-3). Gherkin has natural-languagebased syntax and it is used to describe various scenarios combined. Each scenario can be described by sequence of steps. Each step can be either precondition, action or result. Set of logically combined scenarios is called *feature*. Scenarios collected inside on feature can share common background – list of steps executed before every scenario. Example scenario, testing basic behaviour of chain network, is available in listing [3.21.](#page-59-1)

Listing 3.21: Feature describing basic behaviour of chain network

<span id="page-59-1"></span>

Feature: Testing basic chain network	
Background: 7 nodes network in perfect environment	
Given simulation of 02_chain network in perfect environment	
And every router is alive	
And topology is discovered	

```
Scenario: Send from first node to second node
  When 1 sends to 2 message "hello"
  Then 2 receives from 1 message "hello"
Scenario: Send from first node to last node
  When 1 sends to 7 message "hello"
  Then 7 receives from 1 message "hello"
  And 1 receives acknowledge from 2
  And 2 receives acknowledge from 3
  And 3 receives acknowledge from 4
  And 4 receives acknowledge from 5
  And 5 receives acknowledge from 6
  And 6 receives acknowledge from 7
```
Variety of steps are provided in order to efficiently and accurately write scenarios. List of designed steps is available in table [3.17.](#page-61-0)

With possibilities given by implemented steps, comprehensive feature descriptions has been written. These features covers variety of network functionality such as topology discovery, packet routing, handling undelivered messages and behaviour in variety of environments. Behaviour tests made whole process of protocols' implementation much easier as every aspect of network behaviour was tested for proper operation.

<span id="page-61-0"></span>



# **Chapter 4**

# **Sample applications**

This chapter describes sample applications created in order to present the usage of designed protocols. Each application is written in  $C++11$  as a command line program and uses API of network driver described in section [3.2.3.](#page-42-1) Echo and temperature reading applications, beside driver API, use only standard  $C_{++}$  [1](#page-62-0)1 library. Console application use GNU Readline<sup>1</sup> library in addition.

# <span id="page-62-2"></span>**4.1 Echo**

<span id="page-62-1"></span>**#include** "../driver/driver.h"

Echo is simple application that shows basics of receiving and sending data using designed network. Purpose of this application is to receive data and send back exactly the same data to the sender of received data.

Echo as an optional parameter takes a port number on which it accepts data packets. By default it listens on the port 15. After determining port number, application creates instance of PUT::CS::XbeeRouting::Driver class and invokes listen method on this instance. Listing [4.1](#page-62-1) shows whole implementation of the program.

Listing 4.1: Echo application

```
using namespace PUT::CS::XbeeRouting;
int main(int argc, char const* argv[]) {
 Driver* driver = new Driver();
 uint8_t port;
 if (argc == 1) {
    port = 15;
 } else {
    port = atoi(argv[1]);
 }
 driver->listen(port, [ = ](Address source, uint8_t* data, size_t length) {
    if (source != Driver::SELF)
      driver->deliver(source, port, data, length);
 });
 pause();
  return 0;
}
```
<span id="page-62-0"></span><sup>1</sup>http://cnswww.cns.cwru.edu/php/chet/readline/rltop.html

<span id="page-63-0"></span>

Figure 4.1: Example of using echo application

First argument of listen method is port in which echo application listen for packets. Second argument of listen method is handler for received messages. In this case, it simply sends back received data as new packet using deliver method of PUT::CS::XbeeRouting::Driver instance. The network from figure [4.1](#page-63-0) reveals the following usage examples:

- 1. node *A* sends message to node *C*,
- 2. node *C* reads this message and sends the same data as in received message back,
- 3. node *A* receives message from node *C* with the same data as sent in point 1.

## **4.2 Temperature reading**

Another example of application that uses designed protocols is temperature reading. Application is designed for RaspberryPi [\[UH12\]](#page-88-4) with Raspbian operating system, but it should also work on notebooks and PCs with Debian like Linux distributions. It reads actual temperature of CPU and sends that value on demand. Similar to echo application, it takes one optional argument, which is port number for incoming requests from other nodes. By default, port number equals 7.

Application reads temperature of CPU from suitable system file and divides value by 1000 to obtain temperature in Celsius degree. Listing [4.2](#page-63-1) shows how this is done:

Listing 4.2: Reading temperature of CPU

```
using namespace std;
float temperature() {
  ifstream f("/sys/class/thermal/thermal_zone0/temp");
  if (!f.is_open())
    return -1;
  int temperature = 0;
  f >> temperature;
  return temperature / 1000.0;
}
```
Main functionality of this application is similar to echo. Its code consists of obtaining port number, creating instance of PUT::CS::XbeeRouting::Driver class and invoking listen method on it. Implementation differs only in handling received data. This is shown in listing [4.3:](#page-63-2)

Listing 4.3: Handler for received data

```
driver->listen(port, [ = ](Address source, uint8_t* data, size_t length) {
   if (source != Driver::SELF) {
     if (strncmp((char*)data, "gettemp", 7) == 0) {
        driver->deliver(source, port, (uint8_t*)to_string(temperature()).c_str(), s.length());
     }
   }
 });
```
Handler checks if source node of received data is not self node. Then checks whether data is equal to "gettemp" which is proper request for fetching response from this application. When the above requirements have been meet, the obtained temperature is sent to request source using deliver method on PUT:: CS:: XbeeRouting:: Driver instance.

Example use case for temperature reading application and network topology from figure [4.2](#page-64-0) can be as follows:

- 1. node *A* sends "gettemp" message to node *C*,
- 2. node *C* reads temperature of its CPU,
- 3. node *C* sends this temperature to node *A*,
- 4. node *A* receives message with temperature from node *C*.

<span id="page-64-0"></span>

Figure 4.2: Example of using temperature reading application

## **4.3 Console**

Applications described above react only when data is sent to them. Purpose of console application is to allow users to send data to chosen node at chosen port from command line. This application was very useful as debugging tool while implementing communication protocols.

Console application takes two command line arguments while launching: first one is port number (this is required argument), on which application listen for packets and second one is path to file or device to which information about packets will be written. As default this argument is set to stdout.

Whole application, excluding obtaining command line arguments, consists of two parts. First one creates PUT::CS::XbeeRouting::Driver instance and invokes listen method on this instance. This is shown in listing [4.4:](#page-64-1)

Listing 4.4: Creating Driver instance and invoking listen method

```
Driver* driver = new Driver();
driver->listen(port, [ = ](Address destination, uint8 t p, Address source, uint8 t* data, size t
         length) {
   if (source == Driver::SELF)
        fprint(descriptor, "packet<sub>u</sub>send:\n'tot: <math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<math>u</math>,<mathdestination, p, length, data);
   else
        {\tt fprint((descriptor, "packet_Ureceived:\\ \verb|http://d,lport:_0{d,l.length:_0{zul\verb|drat:_0{shln",source, p, length, data);
});
```
First argument of listen method is port in which console application listen for packets. The second one is handler for incoming and outgoing packets. It simply writes to chosen file/device information about received/sent packets and data from these packets.

Second part of application is responsible for displaying command prompt and parsing commands typed by user. Listings [4.5](#page-65-0) and [4.6](#page-65-1) show how this is implemented.

Listing 4.5: Main loop of application

```
bool run = true;
char* input, shell_prompt[100];
while (run) {
  // create prompt
  \texttt{snprint}(shell\_prompt, \texttt{ sizeof}(\texttt{shell\_prompt}), \texttt{ "port:} \% d \texttt{>} \texttt{''}, \texttt{ port});input = readline(shell_prompt);
  if (!input)
    break;
  if (strcmp(input, "") != 0)
    add_history(input);
  if (strcmp(input, "exit") == 0)
    run = false;
  else
    parse_command(input, driver, port);
  free(input);
}
```
Listing [4.5](#page-65-0) shows main loop of application. It simply creates command prompt and displays it using readline function. This function also reads input typed by user and stores it in input variable. Then, if input variable is not null and is not an empty string, typed command is added to command history using add\_history function, so it is accessible using up and down arrows (like in bash). readline and add\_history functions come from GNU Readline Library. Next, it checks if command is not equal to "exit", which results exiting application. If it is not, then parse\_command function, shown in listing [4.6,](#page-65-1) is invoked.

Listing 4.6: parse command function

```
void parse_command(char* command, Driver* driver, uint8_t port) {
  char* action = (char*)malloc(sizeof(char) * 10);
  int destination = 0;
  char* data = (char*)malloc(sizeof(char) * 255);
  \texttt{sscanf}(\texttt{command}, \text{''s_u\&d_u\&[\^n]s", \texttt{action}, \&\texttt{destination}, \texttt{data});if ((\text{strcmp}(\text{action}, "deliver") == 0) || (\text{strcmp}(\text{action}, "d") == 0))driver->deliver(destination, port, (uint8 t*)data, strlen(data));
  }
  free(data);
  free(action);
}
```
This function parses command typed by user and invoke proper action for typed command. Command have following structure: action destination data where the only possible action is deliver (or shortcut:  $d$ ) – sends packet to given node on given port using PUT::CS::XbeeRouting::Driver::deliver method, For example, deliver 2 some\_kind\_of\_data will cause delivery some kind of data to node 2 on port given as command line argument to console

application.

## **4.4 Possible areas of usage**

There are many places where designed protocols can be used and many applications which can be written to use them. As examples, system for measuring time during sport events and sensoric networks are presented below.

## **4.4.1 Sport timing**

As mentioned before, one possible area of usage for designed protocols is measuring time during sport events like marathons, triathlons, ski country and so on. Whole system could look as follows. Along entire marathon track, there are gates that records timestamps for every runner when he passes through gate. At every gate and at arbiter position, there is XBeePro 868 [\[Dig11a\]](#page-86-6) device connected to RaspberryPi [\[UH12\]](#page-88-4). These devices create one network using communication protocols and algorithms described before. At every gate, there are RFID [\[Wan06\]](#page-88-5) aerials with controller and they are connected to RaspberryPi device. Every runner has RFID chip pinned to his leg. When he passes through gate, application on RaspberryPi gets through RFID aerials and controller unique identifier of runner stored in RFID chip, concatenates it with actual timestamp and sends this information to arbiter position via created at the beginning network. Then, at arbiter position, received data are matched with name of runner, stored with identifier in database and can be shown to arbiter and supporters.

## **4.4.2 Sensoric networks**

Another area of usage for designed protocols is measuring different kinds of parameters in pipelines or other transmission networks. For example, protocols can be applied to measuring pressure in gas pipeline. Scenario is very similar to sport timing. Along entire pipeline, there are places where XBeePro 868 [\[Dig11a\]](#page-86-6) connected to embedded system are placed. At control station, there is another XBeePro 868 device connected to computer that process received data. These devices create network described in previous chapters. Every RaspberryPi device along pipeline has connected pressure sensor. Application reads periodically pressure and sends this information to control station, where this information is processed. Obviously, there is nothing against the use of greater number of sensors with single RaspberryPi device, but the number of input/output ports in RaspberryPi and throughput of system should be taken into account.

# **Chapter 5**

# **Performance evaluation**

In order to analyze the protocols, variety of tests are performed. Firstly topology discovery time is investigated, than message routing was benchmarked using load tests, spike tests and harsh environment tests [\[Sub06\]](#page-88-6). Such tests should give back clear image of protocols' design and implementation behaviour. Analysis of protocols' behavior under the load and in prone to crashes environment is accomplished in the section [5.6](#page-79-0) at the end of this chapter.

# **5.1 Methodology**

All of tests have been carried out with simulator (subsection [3.2.6\)](#page-51-0) to provide the required test conditions and avoid random interference. Moreover the use of a simulator provided tests repeatability. After measuring XBeePro-868 devices' transmission delay, delays on each link were simulated with normal distribution determined by 20ms mean and 1ms standard deviation. All other parameters that do not have direct influence upon test behaviour, such as message contents, have been generated randomly.

Message roundtrip time was chosen as benchmark measure. Therefore, each node was running *echo* application (section [4.1\)](#page-62-2). Time measured between sending message from source node to destination and receiving it back from destination on source node is reliable indicator of protocols behavior. With such response time, protocols' and implementations' performance can be easily compared in various kinds of network topologies. Furthermore as short as possible roundtrip time is a desirable from the end-user perspective, because request-response is common network application workflow. Each test was performed 20 times and the reported results are the average values. Complete numerical values can be found in Appendix [A.](#page-82-0)

# **5.1.1 Tests environment**

Significant tests parameters are network topology and its size. Following topologies were selected for every test:

- chain graph every node is connected to maximum two nodes and there is no cycles (figure [5.1\)](#page-69-0),
- spider web graph topology resulting of star and ring union (figure [5.1\)](#page-69-0),
- random sparse graph random graph generated using Gilbert model [\[Gil59\]](#page-87-4) with 0.02 to 0.2 probability of edge existence, depending on network size (figures [5.3,](#page-70-0) [5.4\)](#page-70-1).

<span id="page-69-0"></span>

Figure 5.1: Example chain network visualization (5 nodes)



Figure 5.2: Example spider network visualization (10 nodes)

It is worth noticing that chain graph is a pessimistic network topology, so it holds the upper bound of roundtrip time, it is least reliable kind of topology, however it offers the greatest geographical span. Spider web graph can represent real world network built with radio modules with omnidirectional antennas arranged around one central node. However the most important topology is random sparse graph – it represents real world network, where every node has between two and four direct neighbours and nodes positions are distributed quite unpredictably.

For each topology, following order of graph sizes values *n* were used: 5, 10, 20, 30, 40, 50 to examine behaviour in variety of network sizes. Other parameters like number of sent messages will be individual for each tests and are presented in appropriate sections.

## **5.1.2 Results visualization**

In order to facilitate the analysis of the collected results, test results are visualized on the line charts in appropriate sections. In each test case there is one line chart to the one topology. Charts present average response time dependence on the number of nodes. If test assumes variability of parameter *m* (number of messages), then on each chart suitable number of curves is presented.

Test results should confirm or refute the claims, that protocols work correctly on different conditions. The shape of the curves determine the behaviour of protocols. It is expected that with the increased order of graph the response time will also rise, as response time between furthest nodes is dependent on network diameter. However, this growth should not be exponential. Moreover curves for lower values of parameter *m* should be dominated by curves for greater value of this parameter.

<span id="page-70-0"></span>

Figure 5.3: Example sparse random network visualization (10 nodes)

<span id="page-70-1"></span>

Figure 5.4: Example sparse random network visualization (40 nodes)

<span id="page-71-0"></span>

Figure 5.5: Complete topology discovery time

# **5.2 Topology discovery**

Important feature of the presented protocol is self-organization, an essential part of this feature is topology discovery. It is limited by sending out a broadcast in wireless network. Multi-hop nature makes it only more difficult and time-consuming. In order to examine how the protocols handle topology discovery, the following test was conducted.

In the same moment all nodes become enabled (their state is changed from passive to active) and they join the network, this moment of time will be referred to as start time. Moment when last node discovers complete network topology (that is every node achieved consistent and final state of network topology) is called discovery end time. Difference between these two times, discovery end time and start time, is interpreted as complete topology discovery time.

It is assumed that all nodes initiate topology discovery protocol at the same time. This behaviour allows discovering various parts of the network at the same time. However, some mutual interference effects can be observed with adjacent nodes.

In figure [5.5,](#page-71-0) it can be seen that chain topology discovery is slowest as each node is connected to one or two other nodes only. That fact causes chain network to be the worst case, yet still it can be noticed that complexity is linear in network size domain.

In case of spider web topology, observed outcome seems to be optimal. Increasing the network's size results in slight increase of topology detection time and the curve is resembling logarithmic complexity. The fact of quick reach of a consistent state results from small network diameter and constant degree for all vertices (excluding the central node). Detection of a new node can be propagated to every other node in a very small time, as the central node is connected to every node. These factors make this topology an optimal one.

Topology discovery in random sparse graph seems to be little worse than in spider web topology.
It can be seen, that their curves are closely related. As random graph reflects real world scenario, measured 750ms required to discovery topology on every node seems promising.

It is worth mentioning that after network topology is discovered (the consistent state is reached), new node joining the network will obtain complete network graph in constant time. As only one heartbeat and graph transmission from any adjacent node is required to acquire network graph. this operation can be completed in less than 50ms. However, propagation of this new node in to whole network is bound by the laws presented before.

#### **5.3 Load testing**

Load testing is measuring network behaviour when it is under load. In case of this test it is necessary to simulate load. To realize this task, 20 messages are continuously sent between random nodes.

Test system sends *m* number of special messages from node *A* to node *B* and measures the roundtrip time. Where *A* and node *B* nodes are the outermost vertices on the graph diameter. Eventually time between sending first of *m* messages and receiving back the last message is used as measure. Presented numbers of special messages (*m*) are used: 5, 10, 20, 30, 40, 50.

Chain topology chart (figure [5.6\)](#page-72-0) shows linear response time rise with increasing number of nodes. As chain topology is worst case scenario, it strongly dominates other topologies' behaviour.

In spider web topology load test (figure [5.6\)](#page-72-0), it can be noticed that roundtrip time is not dependent upon order of the graph. This is correct behaviour due to to the fact, that in average case at most one intermediate node is required to connect pair of nodes. In consequence response time is more less constant.

<span id="page-72-0"></span>

Figure 5.6: Load test in chain network



Figure 5.7: Load test in spider network

<span id="page-73-0"></span>

Figure 5.8: Load test in sparse random network

<span id="page-74-0"></span>

Figure 5.9: Spike test to random node on chain network

Results for random sparse graph (figure [5.8\)](#page-73-0) are the most interesting as this topology reflects real world situations. It can be seen, that the curves rise a peak at 30 nodes – this phenomenon is closely correlated with graph diameter. The graph diameter grows with the order of graph, until some threshold number of nodes is reached, where the diameter starts decreasing. This is a result of the chosen graph generation method.

#### **5.4 Spike testing**

Spike testing is measuring system response when system load suddenly increased.

Two useful varieties of this test were performed:

- messages are sent from each node to other randomly chosen node (figures [5.9,](#page-74-0) [5.10,](#page-75-0) [5.11\)](#page-75-1),
- messages are sent from each node to their furthest node (figures [5.12,](#page-76-0) [5.13,](#page-76-1) [5.14\)](#page-77-0).

It should be noticed that transfers occur concurrently, that is each node starts sending *m* messages independently in the same time, so in total  $m \cdot n$  messages are transferred. Tests were performed for following number *m* of messages: 5, 10, 15, 20, 30.

Measured roundtrip times are lower in comparison with load tests results, even though multiple transfers occur in parallel. This is a consequence of a smaller number of messages circulating in the network. Spider web graph still is leading network topology, no matter if transfers occur between random or furthest nodes. Observed behaviour of network based on random sparse graph is acceptable as roundtrip times are far better than values of chain network, which is a pessimistic topology.

<span id="page-75-0"></span>

Figure 5.10: Spike test to random node on spider network

<span id="page-75-1"></span>

Figure 5.11: Spike test to random node on random network

<span id="page-76-0"></span>

Figure 5.12: Spike test to furthest node on chain network

<span id="page-76-1"></span>

Figure 5.13: Spike test to furthest node on spider network

<span id="page-77-0"></span>

Figure 5.14: Spike test to furthest node on random network

Conclusively, it can be noted that concurrent messages in network do not cause unpredictable behaviour.

#### **5.5 Harsh environment testing**

Harsh environment is defined as network, where randomly occurring failures of nodes or retransmissions can happen. In real life situation network effective topology can change as a result of bad weather, power failures or interferences.

With simulator implemented in the thesis it is possible to mimic such environment. Therefore it is possible to test protocols and their implementation easily in controlled conditions.

The harsh environment test is based on load test, however different environment is assumed:

- a) constant 50% probability of retransmission on every edge (figure [5.15\)](#page-78-0),
- b) constant 75% probability of retransmission on every edge (figure [5.16\)](#page-78-1),
- c) 40% probability of retransmission on every edge 100ms after start (figure [5.17\)](#page-79-0).

As harsh environment test is closely related to real world situation, tests were conducted only on random sparse graphs. Known number *m* of messages (where *m* is 5, 10, 15, 20) are sent from random node *A* to its furthest node and roundtrip time is measured. Time required to receive back every sent message is used as measure.

Harsh environment testing is fundamental in estimating protocol's stability in prone to crashes real world environment.

<span id="page-78-0"></span>

Figure 5.15: Harsh environment testing with  $50\%$  probability of retransmission

<span id="page-78-1"></span>

Figure 5.16: Harsh environment testing with 75% probability of retransmission

<span id="page-79-0"></span>

Figure 5.17: Harsh environment testing with 40% probability of retransmission 100ms after start

In presented charts similar peak as in load tests can be noticed, however its source is the same as previously described. In comparison with load tests [5.8,](#page-73-0) measured roundtrip times have increased considerably, yet still shapes of the curves are comparable. Time increase results from packet retransmissions required to ensure the delivery (during the test no undelivered messages were found). Correct protocols behaviour can be observed, even when environment parameters change in time, as in figure [5.17.](#page-79-0)

#### **5.6 Summary**

Performed tests provide sufficient information about the quality of protocols and their implementation. Tests confirm the protocol's ability to be scalable in context of the order of graph and network diameter. It can be noticed that computation time is minor part compared with transmission time, especially when data rate is 24Kbps.

In fact, the benchmark confirmed that designed protocols are adequate for use in real world situations. Tests based on random sparse graph topology were useful in prediction of behaviour in physical world. Topology can be discovered in satisfactory time. Messages can be transmitted effectively, even when under constant load. Moreover network can adapt to changing environment, as it makes use of redundant links in case of other link failures.

### **Chapter 6**

## **Conclusions**

In this thesis we provided the concept of brand new communication protocols for self organizing multi-hop long distance ad hoc networks. As part of the design process, it was proved that these protocols have *No creation* and *Stubborn delivery* properties. We also proved that having these two properties results in protocols correctness and implementing *Stubborn Point to Point links* interface.

Furthermore, sample implementation of designed protocols with example applications, such as echo, temperature reading and console is provided. Also example hardware platform based on XBeePro-868 and RaspberryPi is set up. As there was limited number of devices, network simulator that allows to simulate various network topologies and environment conditions was developed for the purposes of tests.

Various performance tests were conducted using this simulator. Load tests, spike tests and harsh environment tests shown, that scalability of designed protocols depends on size and topology of the network and matches performance expectations. Furthermore, harsh tests also shown ability of protocols to self-organize/reorganize while reliability of edges change.

In addition to conducted tests, provided implementation in connection with example applications and hardware setup shown that designed protocols can be used in real applications.

#### **6.1 Comparing to existing solutions**

Comparing to existing solutions, such as cellular network, ad hoc networks that use designed protocols have many advantages. Firstly, cellular network is not available everywhere, especially away from cities and roads. It is very important in case of measuring time during sport events. Triathlons, for example, can take place away from city and cellular network range. Thanks to the fact, that designed protocols implement multi-hop ad hoc network, large area may fall within a range of the network. And because range of single XBeePro-868 devices with proper antenna is up to 40 kilometers, check point gates that measures time does not have to be close to each other.

Joining new device to the existing network is very simple in provided implementation and is not more complex than in cellular networks. To connect the device to the network, turning the device on and inputting unique address is sufficient. Because designed protocols implement self organizing ad hoc network, device discovers network topology by itself. Furthermore, thanks to self-organizing property, network is able to react and reorganize if any temporary links disturbances occur.

Another advantage, comparing to cellular network is lack of data transfer charge. Cellular network operators charge for even small data transfer, while sending data through ad hoc network

is free if proper bandwidth is used.

The biggest advantage of provided solution is reliability. Failure of one device does not cause failure of whole network. In cellular, failure of base station can cause unavailability of network on large area.

#### **6.2 Further research**

Although the goal of this thesis has been accomplished, there is couple directions in which further research can follow:

- 1. As mentioned above, joining to the network requires setting unique address for each device. Automatic address configuration protocol is worth of investigation for making joining process even simpler.
- 2. Greater ensurance of message delivery can be investigated. Provided solution tries to ensure, that message will be delivered to destination node, but there is still no certainty about this. Increasing delivery ensurance is worth of researching.
- 3. Because of number of available XBeePro-868 devices was limited, performance evaluation was conducted using simulator. Conducting complex tests with usage of physical devices is worth of investigation.

# **Appendix A**

# **Performance evaluation dataset**

	Topology type					
<b>Nodes</b>	random	spider	chain			
5	183.83	133.89	182.05			
10	380.59	261.67	446.28			
20	575.14	359.77	878.22			
30	665.37	460.84	1 408.42			
40	687.04	554.67	1 862.36			
50	717.91	590.51	2 406.63			

Table A.1: Topology discovery time (time in ms)

Table A.2: Load test in chain network (time in ms)

	Messages count						
<b>Nodes</b>	5	10	20	30	40	$50^{\circ}$	
5	774.91	1 570.78	3 383.16	5 052.23	6 759.78	8 038.79	
10	1 636.25	3 789.29	6 375.33	10 678.00	12 537.07	21 407.66	
20	3 455.85	7 1 2 1 .8 6	15 662.50	23 170.68	32 196.94	38 802.32	
30	5 557.82	12 161.15	24 834.30	37 743.08	52 264.37	55 606.29	
40	7 291.13	14 987.34	33 006.21	54 017.31	59 454.87	79 383.53	
50	9 955.75	18 727.36	40 775.20	64 636.29	85 388.45	96 762.51	

Table A.3: Load test in spider network (time in ms)



	Messages count						
<b>Nodes</b>	5	10	20	30	40	50	
5	464.65	801.40	1 698.52	2 915.97	3 756.20	5 045.64	
10	416.51	847.98	1 736.56	2 420.67	3 661.56	5 548.96	
20	411.74	797.75	1 581.62	3 144.31	3 942.15	5 638.12	
30	371.04	837.08	1 743.69	2 529.69	4 4 29 35	4 977.77	
40	482.86	870.82	1 682.84	2 702.47	3 254.15	4 853.75	
50	424.34	764.85	609.20	2 498.20	4 077.27	4 954.96	

Table A.4: Load test in sparse random network (time in ms)

Table A.5: Spike test to random node on chain network (time in ms)

	Messages count					
<b>Nodes</b>	5	10	15	20	30	
5	470.53	1 902.76	1 572.32	2 587.91	2 966.90	
10	216.00	2 459.97	5 049.13	2 858.97	1 808.64	
20	3 661.92	5 439.98	3 459.62	1 061.80	1 739.96	
30	1 137.80	10 613.00	9 296.79	14 137.83	37 369.08	
40	5 856.01	15 386.77	25 4 25 83	17 319.04	40 030.62	
50	1 932.07	17 596.21	10 402.50	45 859.91	20 765.54	

Table A.6: Spike test to random node on spider network (time in ms)

	Messages count						
<b>Nodes</b>	$\overline{5}$	10	15	20	30		
5	420.88	992.33	1 537.02	1 965.11	3 242.56		
10	452.89	913.76	1 380.87	1 872.89	3 301.98		
20	382.23	778.21	1 370.06	2 108.60	3 398.44		
30	416.91	806.74	1 477.04	1 933.06	3 028.07		
40	426.21	861.81	1 359.06	2 030.23	3 715.26		
50	438.39	856.36	1 112.72	1 901.42	3 180.94		

Table A.7: Spike test to random node on random network (time in ms)

	Messages count						
<b>Nodes</b>	5	10	15	20	30		
$\overline{5}$	435.07	1 019.82	1 611.12	2 439.25	6 110.96		
10	464.72	1 392.40	2 2 2 9 . 3 2	2 555.95	5 995.88		
20	1 271.66	877.52	1 685.63	3 559.32	6 190.59		
30	1 193.38	4 017.76	2 991.32	4 547.24	6 206.21		
40	1 089.23	2 698.51	3 054.69	4 957.76	8 480.96		
50	1 123.25	1 491.02	3 780.12	4 533.02	8 976.86		

Table A.8: Spike test to furthest node on chain network (time in ms)



	Messages count					
<b>Nodes</b>	5	10	15	20	30	
5	428.94	835.11	1 348.06	1 778.93	3 044.48	
10	453.19	878.13	1 440.95	1 976.24	2 9 26.91	
20	444.22	944.12	1 465.98	1 870.90	3 104.64	
30	423.26	969.55	1 452.84	1 975.75	3 435.17	
40	424.99	915.58	1 272.10	2 048.87	3 212.02	
50	445.57	916.48	1 470.66	2 206.34	3 173.43	

Table A.9: Spike test to furthest node on spider network (time in ms)

Table A.10: Spike test to furthest node on random network (time in ms)

			Messages count		
<b>Nodes</b>	5	10	15	20	30
5	485.96	954.92	1 545.93	2 408.73	5 619.27
10	682.99	1 420.72	2 360.47	3 797.08	8 483.34
20	1 816.46	3 265.55	5 234.71	8 324.88	18 406.30
30	1 940.31	4 346.74	6 945.03	9 372.48	22 200.96
40	1 284.77	3 059.96	4 399.58	6 512.52	15 309.61
50	101.56	2 300.72	3 811.33	6 215.49	13 671.11

Table A.11: Harsh environment testing with 50% probability of retransmission (time in ms)

	Messages count						
<b>Nodes</b>	5	10	15	20			
5	728.04	1 428.13	2 1 2 3 .0 5	2 913.48			
10	1 154.69	2 262.72	3 447.43	4 551.22			
20	2 897.69	5 819.81	8 570.71	11 812.35			
30	3 712.79	7 712.22	11 479.36	15 124.80			
40	2 411.70	4 969.35	7 535.47	9 893.62			
50	1 979.51	4 106.48	5 867.48	7 851.98			

Table A.12: Harsh environment testing with 75% probability of retransmission (time in ms)

	Messages count						
<b>Nodes</b>	$\overline{5}$	10	15	20			
5	869.47	1 773.25	2 629.32	3 4 4 3 . 9 2			
10	1 548.50	3 118.16	4 593.50	6 251.12			
20	5 316.66	10 241.07	15 432.29	20 242.58			
30	7 300.23	14 686.32	21 966.39	29 194.77			
40	4 093.62	8 334.08	12 505.87	16 620.05			
50	3 227.85	6 487.17	9 595.44	12 830.99			

Table A.13: Harsh environment testing with 40% probability of retransmission 100ms after start (time in ms)



# **Bibliography**

- [ADS86] Bowen Alpern, Alan J Demers, and Fred B Schneider. Safety without stuttering. *Information Processing Letters*, 23(4):177–180, 1986.
- [Asl15] Aslak Hellesoy and Cucumber developers team. Cucumber making bdd fun. [on-line] <http://cukes.info>, 2015.
- [Bar98] Michael Barbehenn. A note on the complexity of dijkstra's algorithm for graphs with weighted vertices. *IEEE transactions on computers*, 47(2):263–263, 1998.
- [BRG99] Meenakshi Bansal, Rachna Rajput, and Gaurav Gupta. Mobile ad hoc networking (MANET): Routing protocol performance issues and evaluation considerations. *The Internet Society*, 1999.
- [CGR11] Christian Cachin, Rachid Guerraoui, and Luís Rodrigues. *Introduction to reliable and secure distributed programming*. Springer, 2011.
- [CMPC04] Paolo Costa, Matteo Migliavacca, Gian Pietro Picco, and Gianpaolo Cugola. Epidemic algorithms for reliable content-based publish-subscribe: An evaluation. In *Distributed Computing Systems, 2004. Proceedings. 24th International Conference on*, pages 552–561. IEEE, 2004.
- [Dic11] American Heritage Dictionaries. *The American Heritage Dictionary of the English Language, Fifth Edition*. Houghton Mifflin Harcourt Trade, 2011.
- [Dig11a] Digi International Inc. Xbee-pro 868 datasheet. [on-line] [http://www.digi.com/pdf/ds\\_xbeepro868.pdf](http://www.digi.com/pdf/ds_xbeepro868.pdf), 2011.
- [Dig11b] Digi International Inc. Xbib-u-dev reference guide. [on-line] [http://ftp1.digi.com/support/documentation/xbibudev\\_referenceguide.pdf](http://ftp1.digi.com/support/documentation/xbibudev_referenceguide.pdf), 2011.
- [Dig14a] Digi International Inc. Xbee rf modules digi international. [on-line] [http://www.digi.com/](http://www.digi.com/products/wireless-wired-embedded-solutions/zigbee-rf-modules/zigbee-mesh-module/) [products/wireless-wired-embedded-solutions/zigbee-rf-modules/zigbee-mesh-module/](http://www.digi.com/products/wireless-wired-embedded-solutions/zigbee-rf-modules/zigbee-mesh-module/), 2014.
- [Dig14b] Digi International Inc.  $X$ bee $\mathbb{R}/x$ bee-pro $\mathbb{R}/868$  rf modules. [on-line] [http://ftp1.digi.com/support/documentation/90001020\\_E.pdf](http://ftp1.digi.com/support/documentation/90001020_E.pdf), 2014.
- [Dig15] Digi International Inc. About digi international. [on-line] <http://www.digi.com/aboutus/>, 2015.
- [Dij59] Edsger W Dijkstra. A note on two problems in connexion with graphs. *Numerische mathematik*, 1(1):269–271, 1959.
- [DR91] Joan Daemen and Vincent Rijmen. The design of Rijndael: AES the Advanced Encryption Standard. *Journal of Cryptology*, 4(1):3–72, 1991.
- [EFGK03] Patrick Th Eugster, Pascal A Felber, Rachid Guerraoui, and Anne-Marie Kermarrec. The many faces of publish/subscribe. *ACM Computing Surveys (CSUR)*, 35(2):114–131, 2003.
- [Eur09] European Union Parliament. Directive of the european parliament and of the council amending council directive 87/372/eec on the frequency bands to be reserved for the coordinated introduction of public pan-european cellular digital land-based mobile communications in the community. [on-line] [http://www.europarl.europa.eu/sides/getDoc.](http://www.europarl.europa.eu/sides/getDoc.do?pubRef=-//EP//TEXT+REPORT+A6-2009-0276+0+DOC+XML+V0//EN) [do?pubRef=-//EP//TEXT+REPORT+A6-2009-0276+0+DOC+XML+V0//EN](http://www.europarl.europa.eu/sides/getDoc.do?pubRef=-//EP//TEXT+REPORT+A6-2009-0276+0+DOC+XML+V0//EN), 2009.
- [Eur11] European Telecommunications Standards Institute. Electromagnetic compatibility and radio spectrum matters (erm); radio frequency identification equipment operating in the band 865 mhz to 868 mhz with power levels up to 2 w;. [on-line] [http://www.etsi.org/deliver/etsi\\_](http://www.etsi.org/deliver/etsi_en/302200_302299/30220801/01.04.01_40/en_30220801v010401o.pdf) [en/302200\\_302299/30220801/01.04.01\\_40/en\\_30220801v010401o.pdf](http://www.etsi.org/deliver/etsi_en/302200_302299/30220801/01.04.01_40/en_30220801v010401o.pdf), 2011.
- [Fac15] Antenna Factor. Ant-868-cw-hwr-rps datasheet specifications: Frequency: 868mhz (853mhz 883mhz). [on-line] <http://www.digchip.com/datasheets/parts/datasheet/1359/ANT-868-CW-HWR-RPS.php>, 2015.
- [Gil59] Edgar N Gilbert. Random graphs. *The Annals of Mathematical Statistics*, pages 1141–1144, 1959.
- [Goo15] Google developers team. googletest, google c++ testing framework. [on-line] <https://code.google.com/p/googletest/wiki/Primer>, 2015.
- [HBC01] Jean-Pierre Hubaux, Levente Buttyán, and Srdan Capkun. The quest for security in mobile ad hoc networks. In *Proceedings of the 2nd ACM international symposium on Mobile ad hoc networking & computing*, pages 146–155. ACM, 2001.
- [JJV08] Jorma Jormakka, Henryka Jormakka, and Janne Väre. A lightweight management system for a military ad hoc network. In *Information Networking. Towards Ubiquitous Networking and Services*, pages 533–543. Springer, 2008.
- [JS03] Jangeun Jun and Mihail L Sichitiu. The nominal capacity of wireless mesh networks. *Wireless Communications, IEEE*, 10(5):8–14, 2003.
- [Lam77] Leslie Lamport. Proving the correctness of multiprocess programs. *Software Engineering, IEEE Transactions on*, (2):125–143, 1977.
- [LK01] Hyojun Lim and Chongkwon Kim. Flooding in wireless ad hoc networks. *Computer Communications*, 24(3):353–363, 2001.
- [Lon14] Roy Longbottom. Roy longbottom's raspberry pi benchmarks. [on-line] <http://www.roylongbottom.org.uk/Raspberry%20Pi%20Benchmarks.htm>, 2014.
- [Per08] Charles E Perkins. *Ad hoc networking*. Addison-Wesley Professional, 2008.
- [PM14] R. Pressman and B. Maxim. *Software Engineering: A Practitioner's Approach*. McGraw-Hill Education, 2014.
- [Poo00] Robert D Poor. Self-organizing network, February 22 2000. US Patent 6,028,857.
- [Ras15] Raspberry Pi Foundation. What is a raspberry pi? [on-line] <http://www.raspberrypi.org/help/what-is-a-raspberry-pi/>, 2015.
- [Red15] Redis developers team. Pub/sub redis. [on-line] <http://redis.io/topics/pubsub>, 2015.
- [RTB<sup>+</sup>13] Daniel G Reina, Sergio L Toral, Federico Barrero, Nik Bessis, and Eleana Asimakopoulou. The Role of Ad Hoc Networks in the Internet of Things: A Case Scenario for Smart Environments. In *Internet of Things and Inter-cooperative Computational Technologies for Collective Intelligence*, pages 89–113. Springer, 2013.
- [Rub13] Ruby development team. pack (array) apidock. [on-line] <http://apidock.com/ruby/Array/pack>, 2013.
- [SNS03] Keng Siau, Fiona Nah, and Hong Sheng. Values of mobile applications to end-users. *European Research Consortium for Informatics and Mathematics (ERCIM) News*, 54:50–1, 2003.
- [Spa13] SparkFun Electronics. Sparkfun xbee explorer dongle wrl-11697 sparkfun electronics. [on-line] <https://www.sparkfun.com/products/11697>, 2013.
- [SS83] Richard D Schlichting and Fred B Schneider. Fail-stop processors: an approach to designing fault-tolerant computing systems. *ACM Transactions on Computer Systems (TOCS)*, 1(3):222–238, 1983.
- [Sub06] B. M. Subraya. *Integrated Approach to Web Performance Testing: A Practitioner's Guide*. IGI Global, 2006.
- [Tij12] Henk Tijms. *Understanding probability*. Cambridge University Press, 2012.
- [UH12] Eben Upton and Gareth Halfacree. *Meet the Raspberry Pi*. John Wiley & Sons, 2012.
- [VDKV00] Arie Van Deursen, Paul Klint, and Joost Visser. Domain-specific languages: An annotated bibliography. *Sigplan Notices*, 35(6):26–36, 2000.
- [Wan06] Roy Want. An introduction to rfid technology. *Pervasive Computing, IEEE*, 5(1):25–33, 2006.
- [YAM06] YAML.org. Yaml ain't markup language. [on-line] <http://www.yaml.org/refcard.html>, 2006.
- [Ye13] W. Ye. *Instant Cucumber BDD How-to*. Packt Publishing, 2013.



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Poznan University of Technology Faculty of Computer Science Institute of Computer Science

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